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(54) **AUTOMATED CUSTOMIZATION OF LOUDSPEAKERS**

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H04R 25/00 (2006.01)
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CPC **H04R 1/00** (2013.01); **G05B 19/4099** (2013.01); **G06F 17/50** (2013.01); **G10K 11/025** (2013.01); **H04R 1/30** (2013.01);
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(58) **Field of Classification Search**

CPC H04R 1/30; H04R 1/20; G10K 11/025
USPC 381/150, 117, 82, 182, 335, 337-342
See application file for complete search history.

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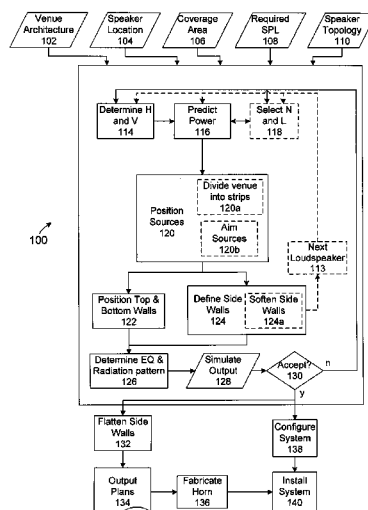
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(57) **ABSTRACT**

In general, in some aspects, a loudspeaker includes a horn including a first end panel, a second end panel, a first side panel, and a second side panel. Edges of at least the first and second side panels define a diffraction slot opening. A plurality of manifold components each has an output opening coupled to the diffraction slot opening. The output openings of the plurality of manifold components together constitute a diffraction slot source at the diffraction slot opening. A plurality of electro-acoustic transducers are each coupled to an input opening of one of the manifold components.

13 Claims, 18 Drawing Sheets



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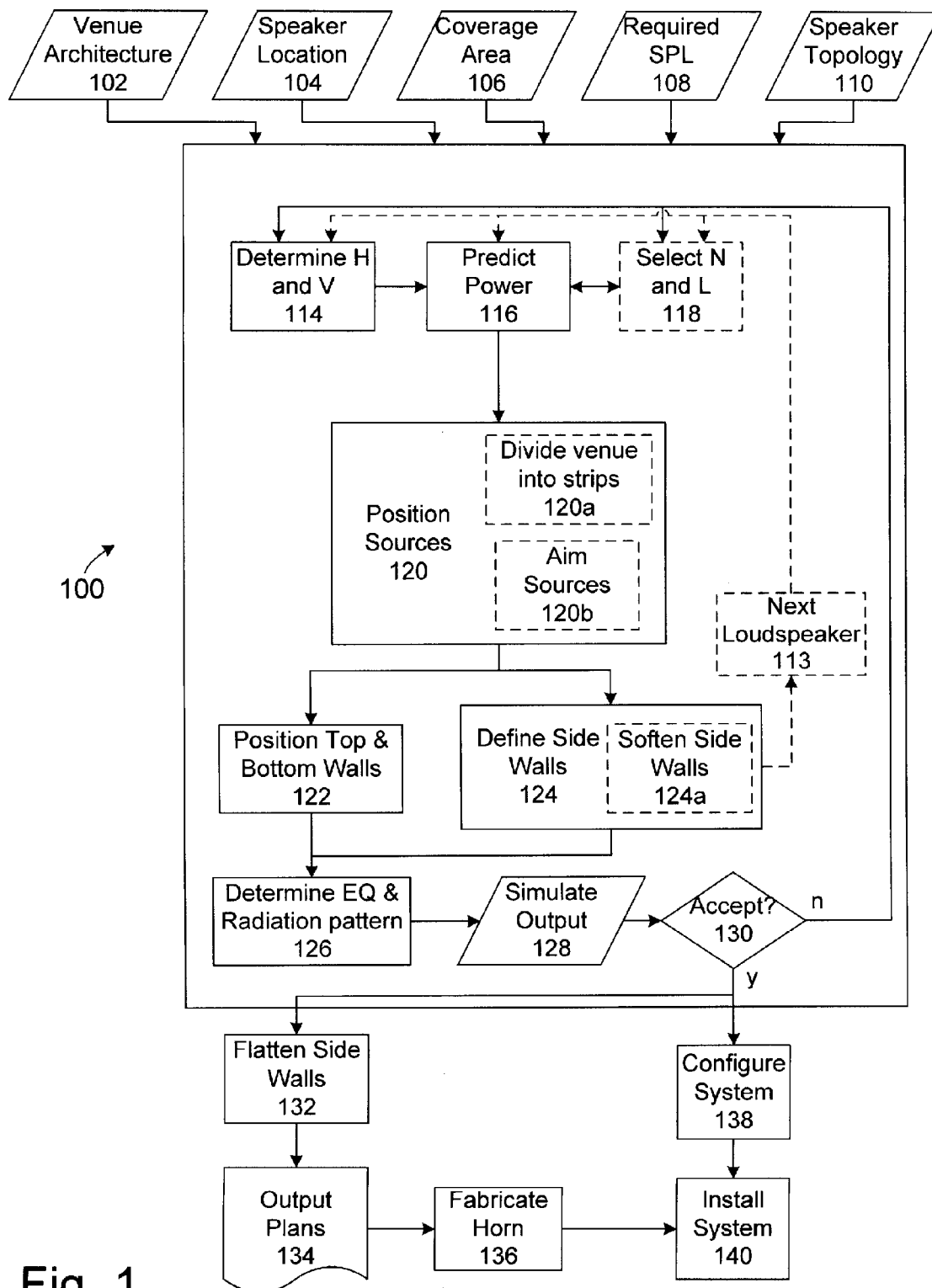


Fig. 1

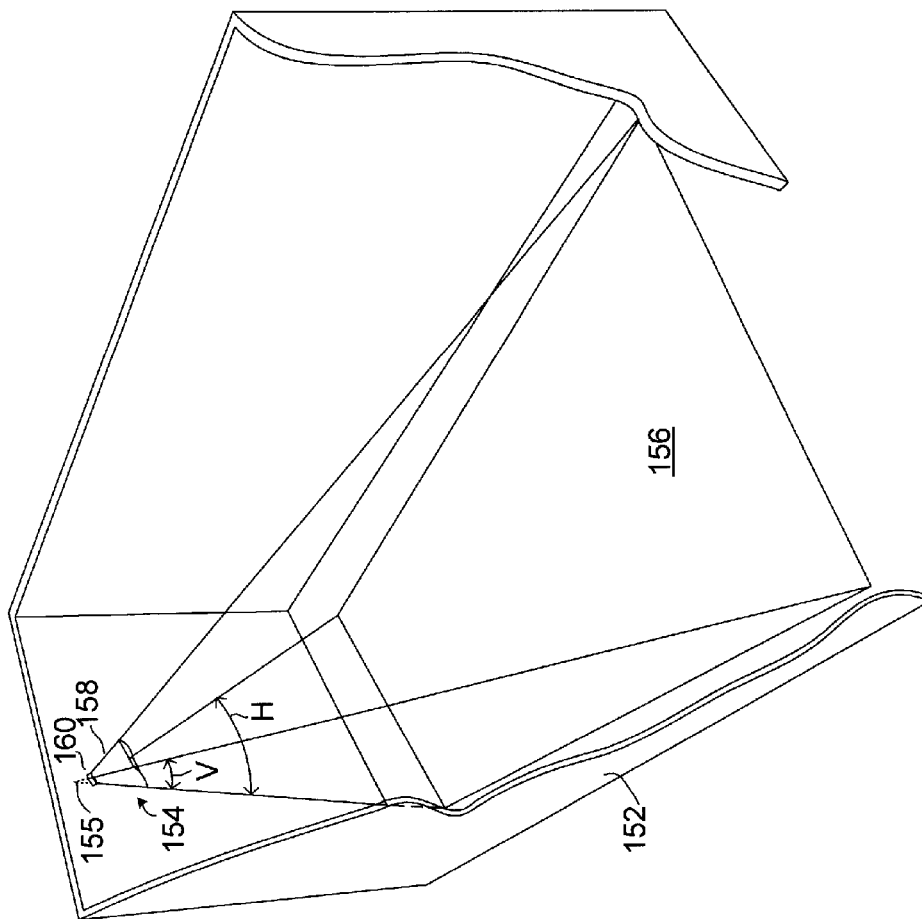


Fig. 2

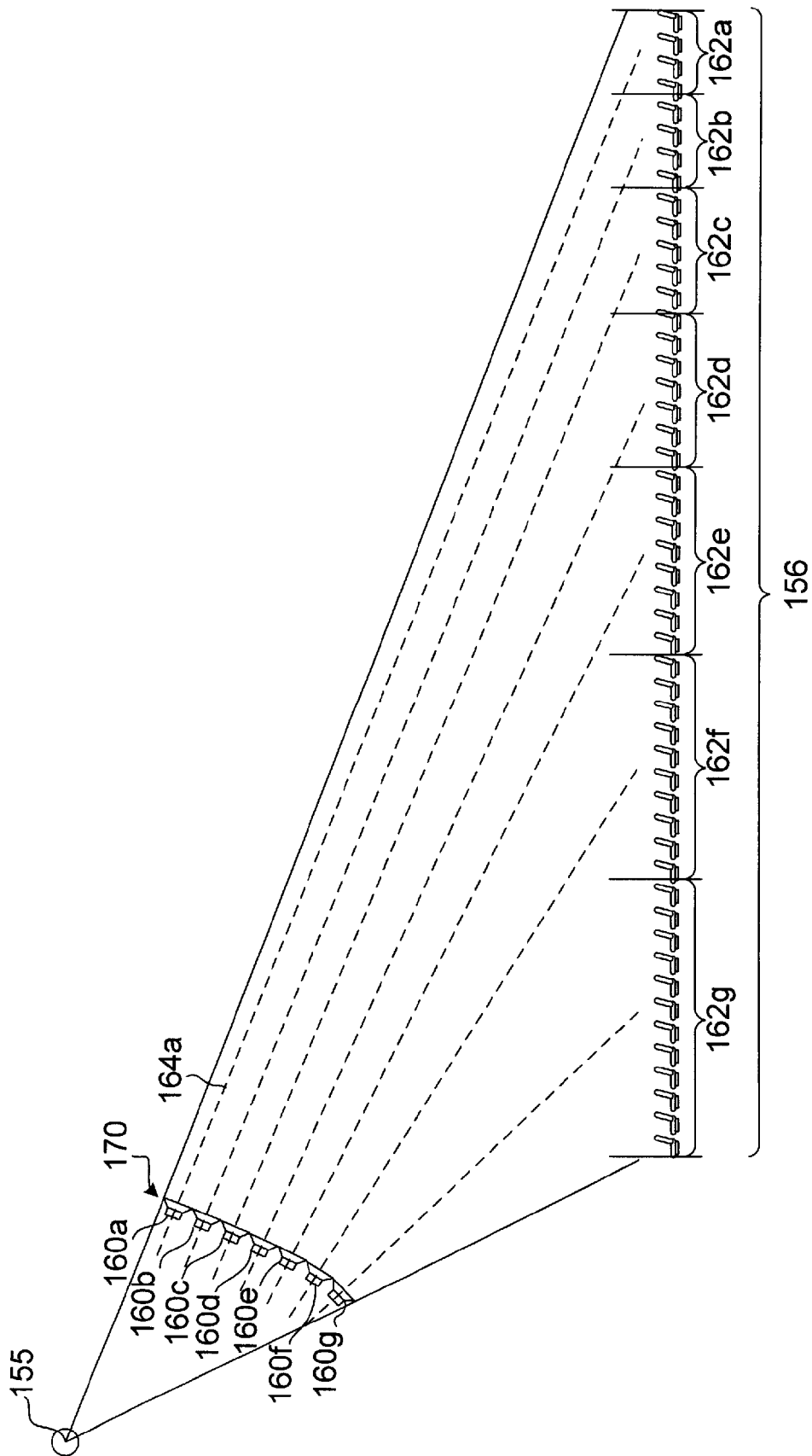


Fig. 3A

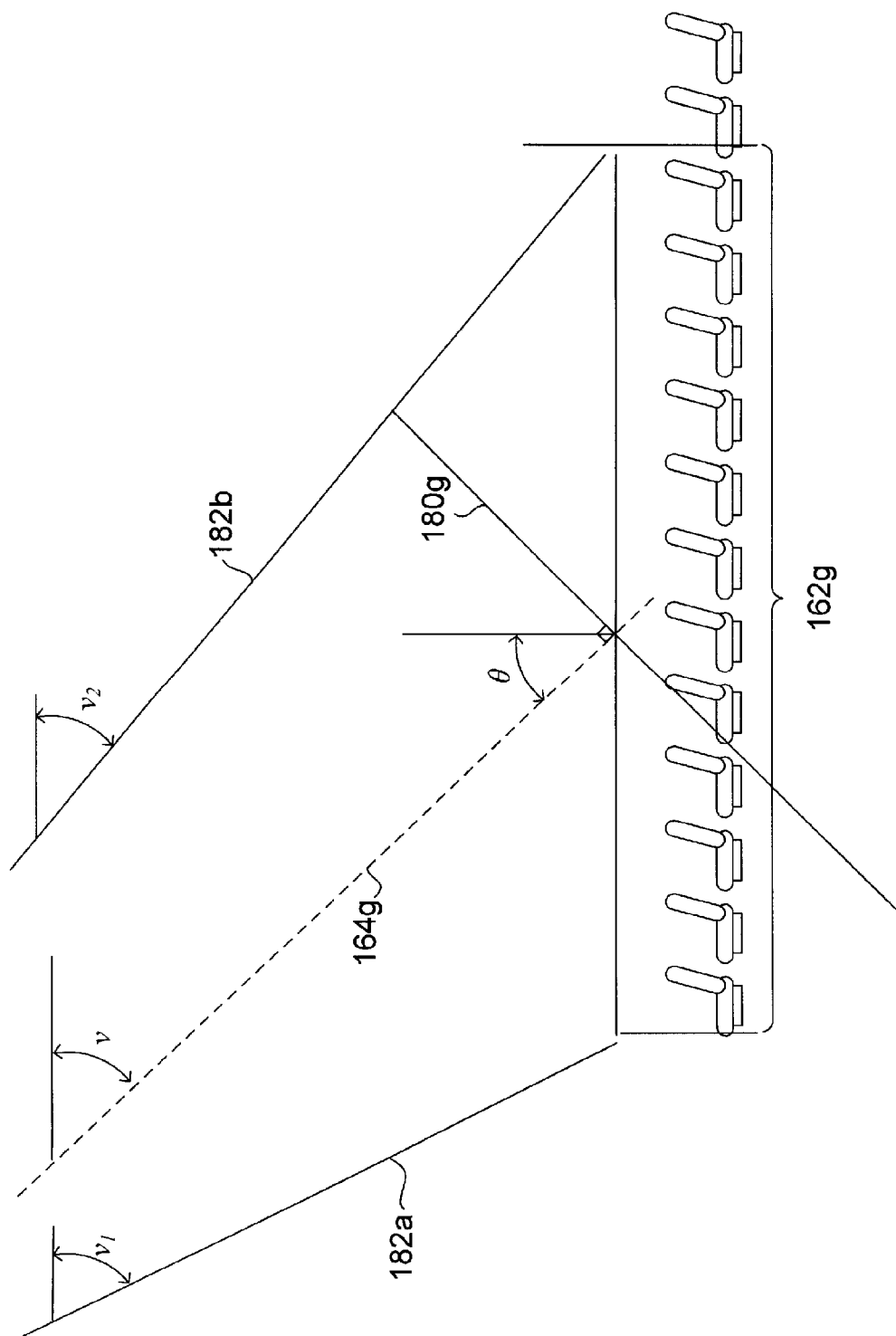


Fig. 3B

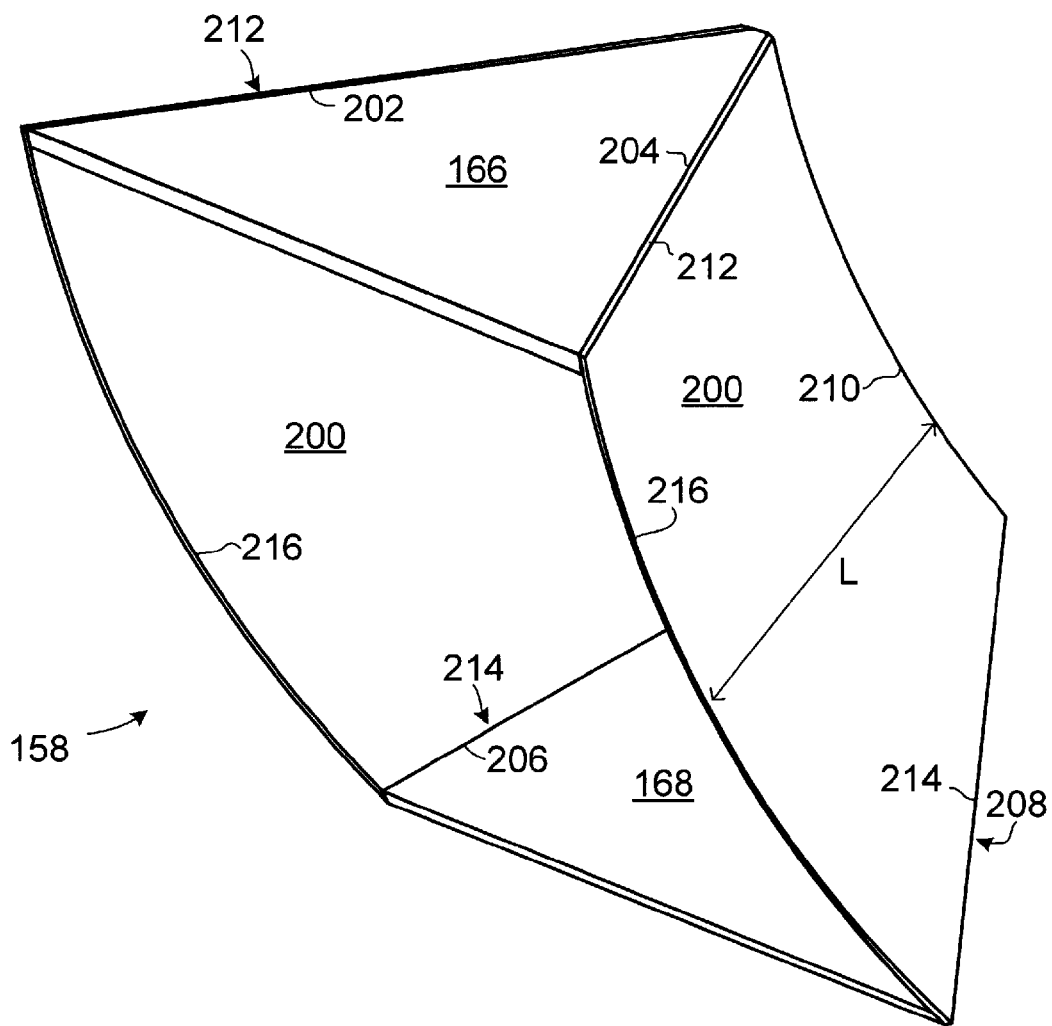


Fig. 4A

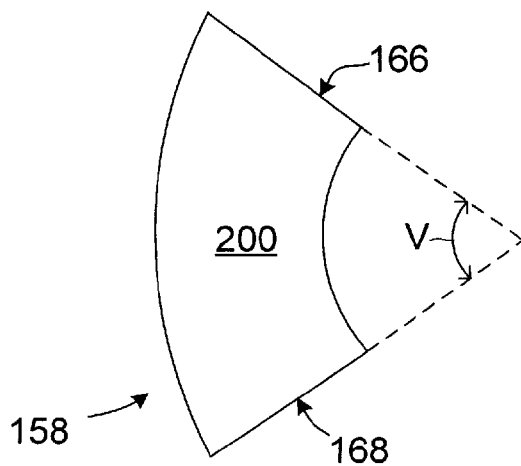


Fig. 4B

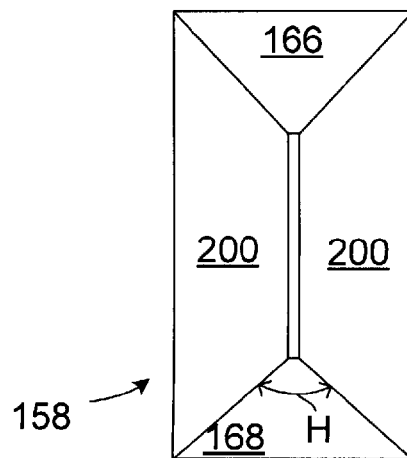


Fig. 4C

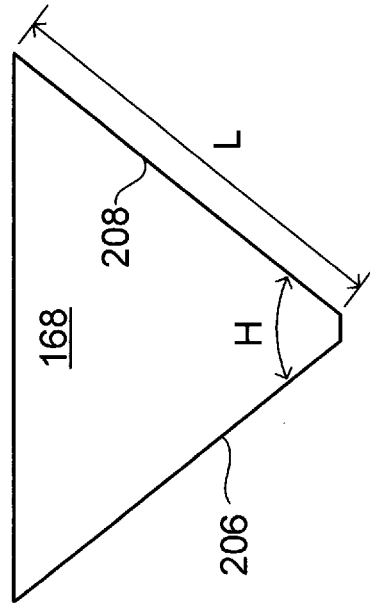


Fig. 5B

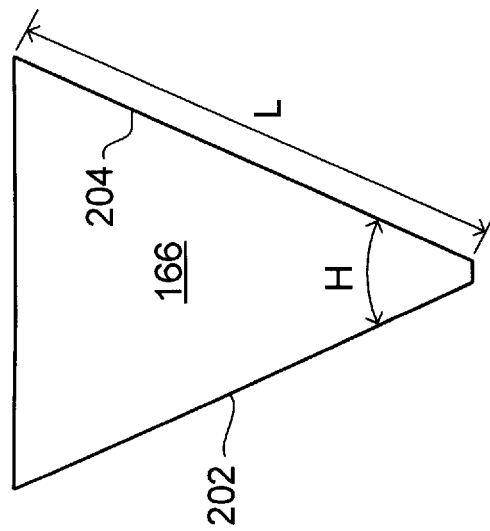


Fig. 5A

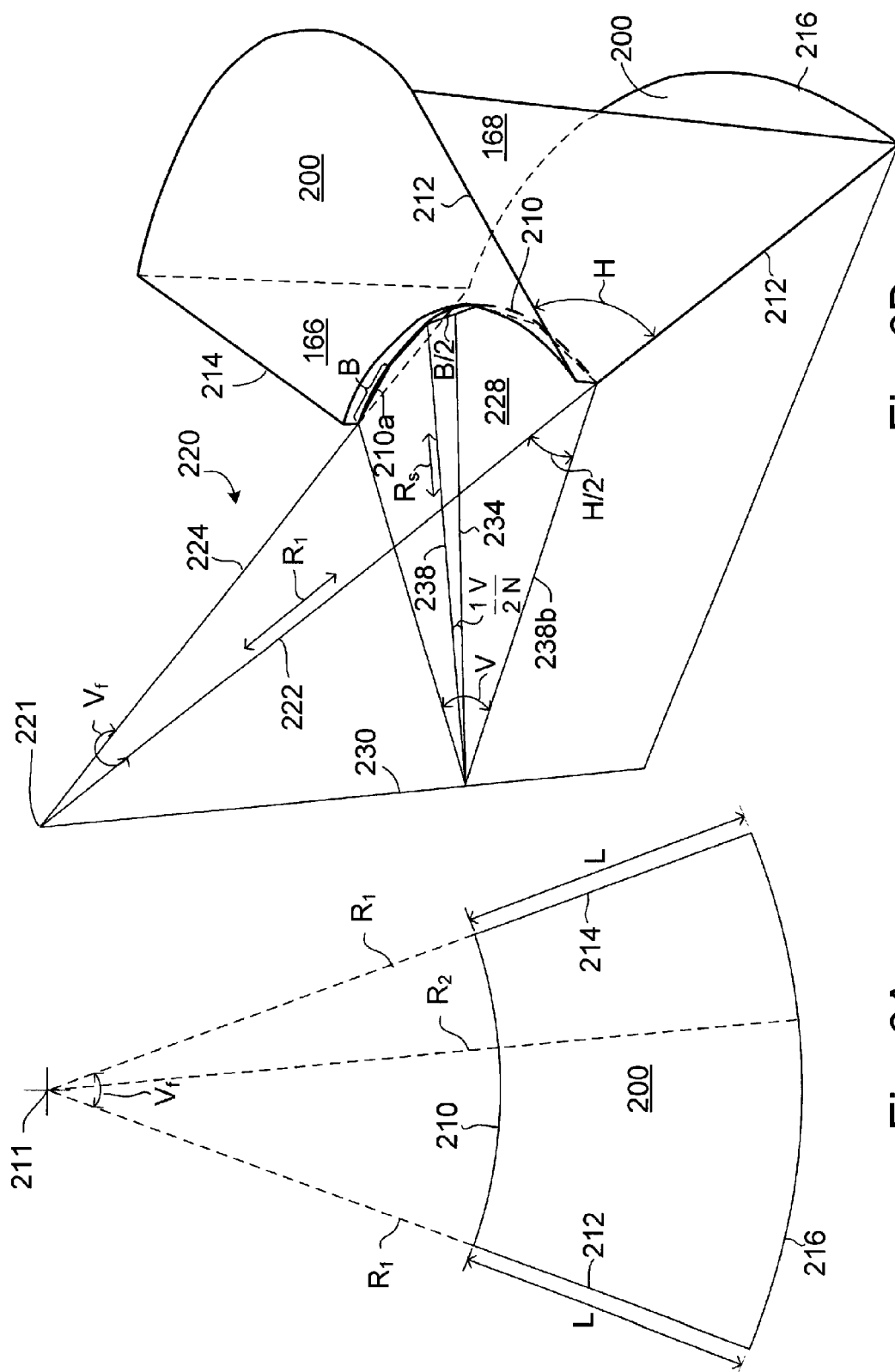


Fig. 6A

Fig. 6B

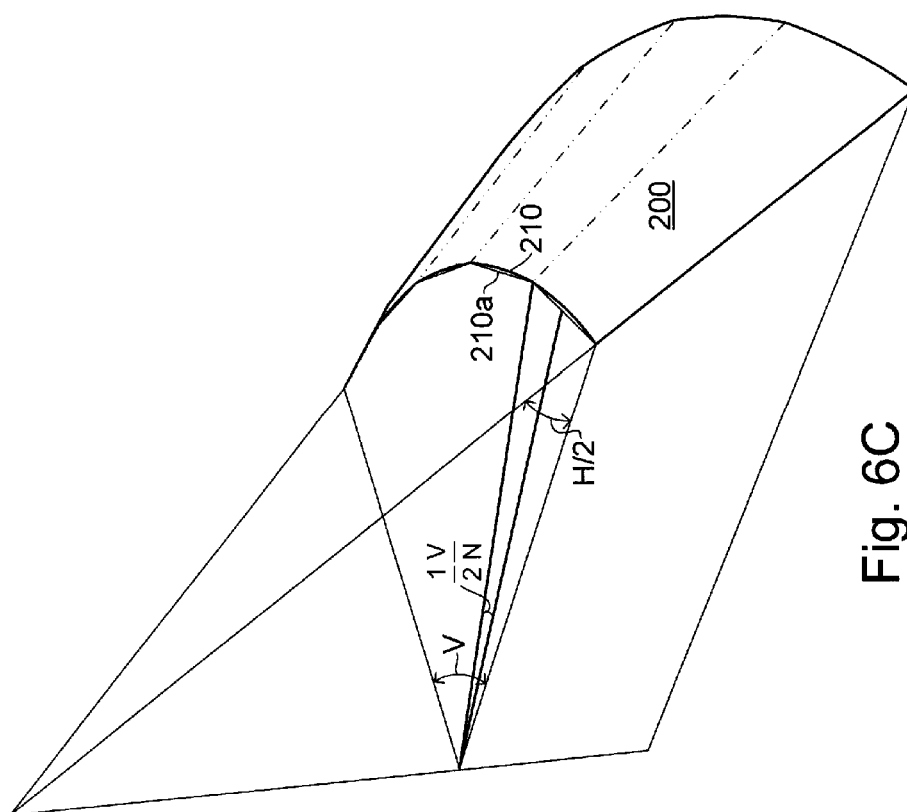


Fig. 6C

Fig. 7B

Fig. 7C

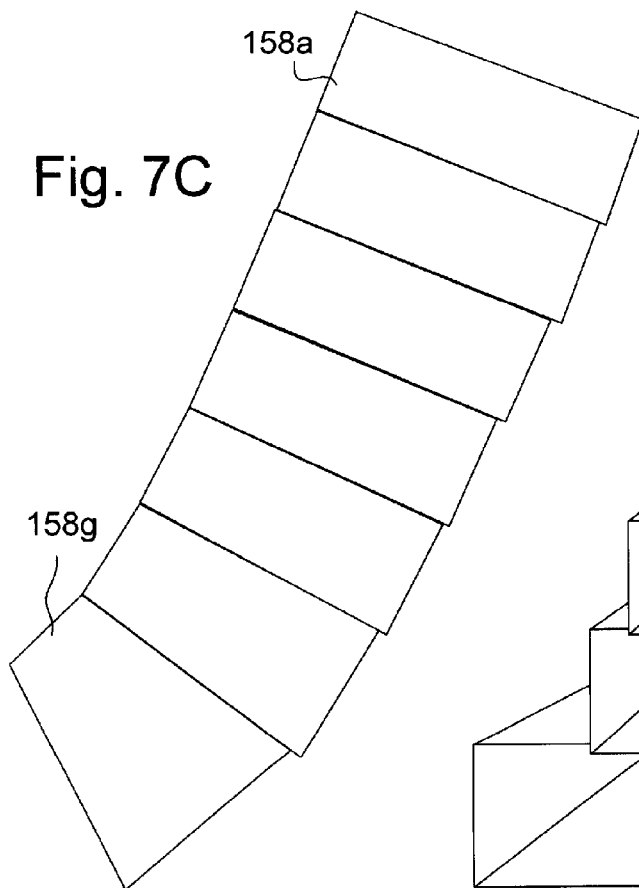
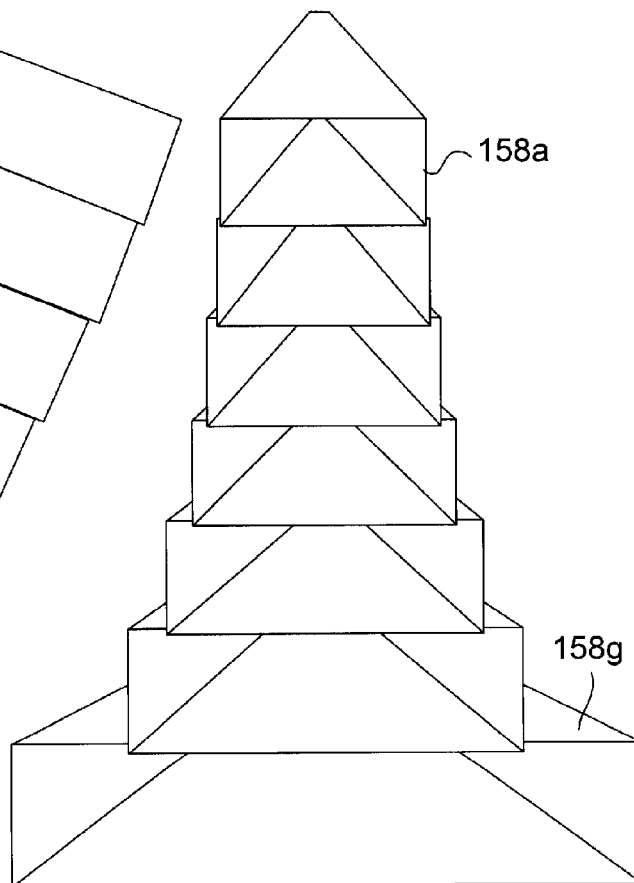


Fig. 7D



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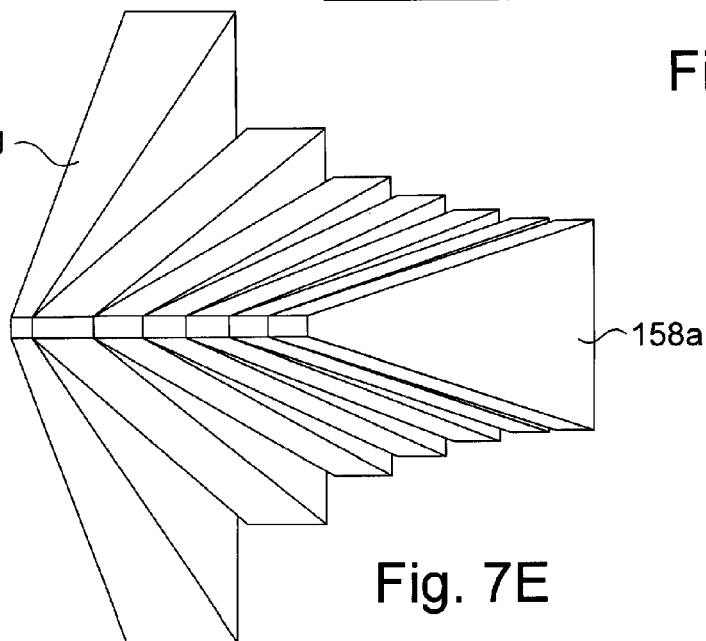
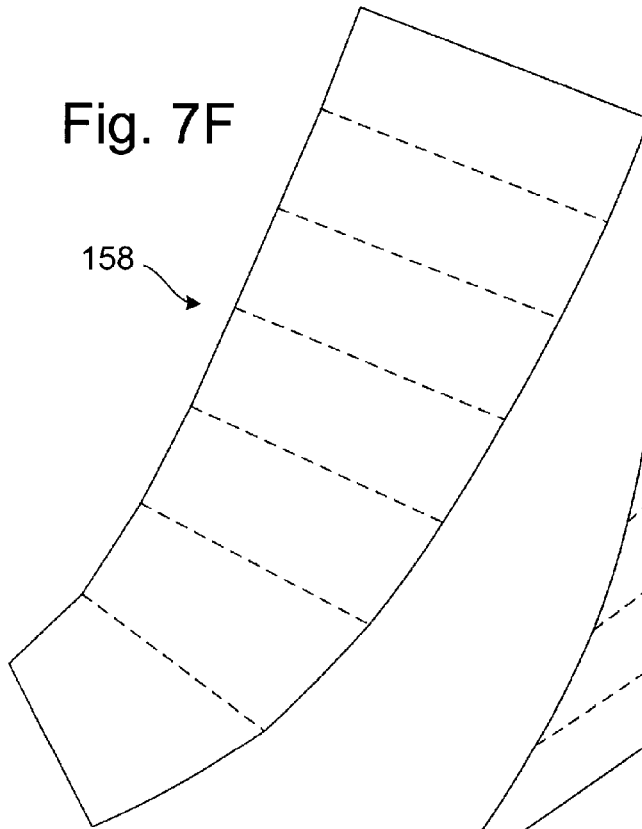


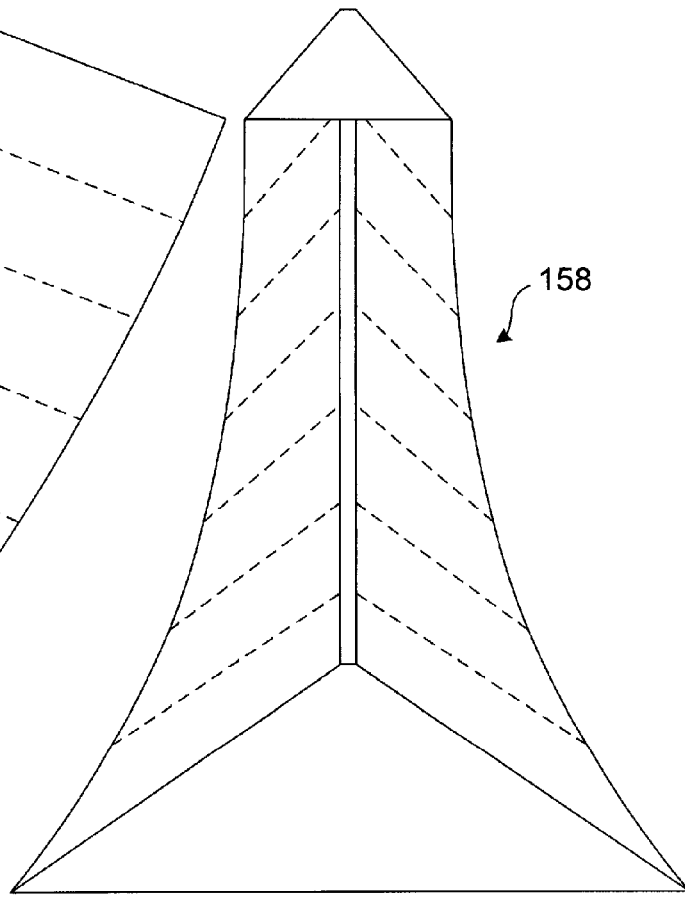
Fig. 7E

Fig. 7F



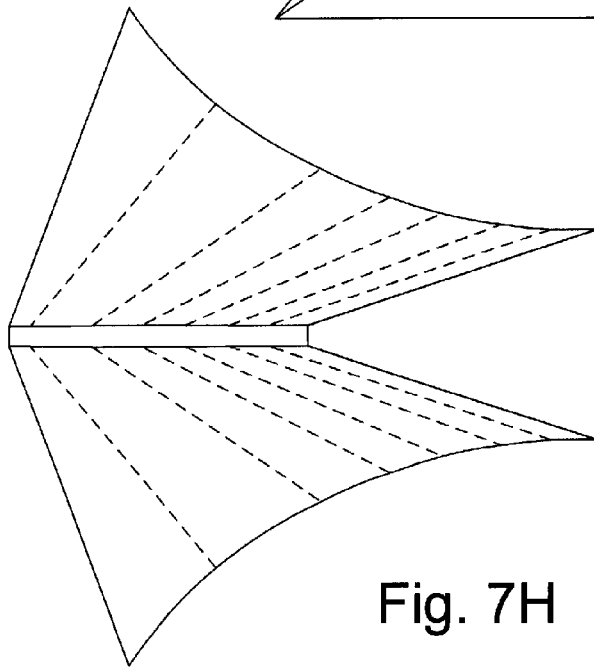
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Fig. 7G



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Fig. 7H



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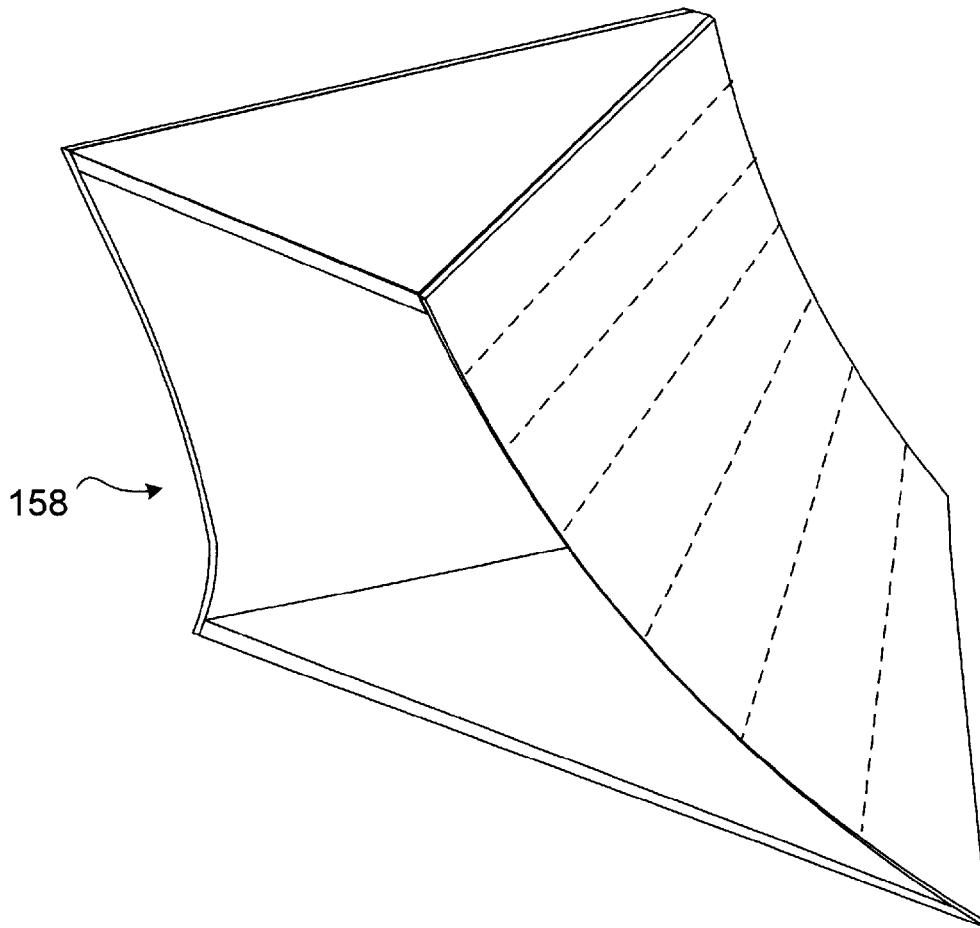


Fig. 7I

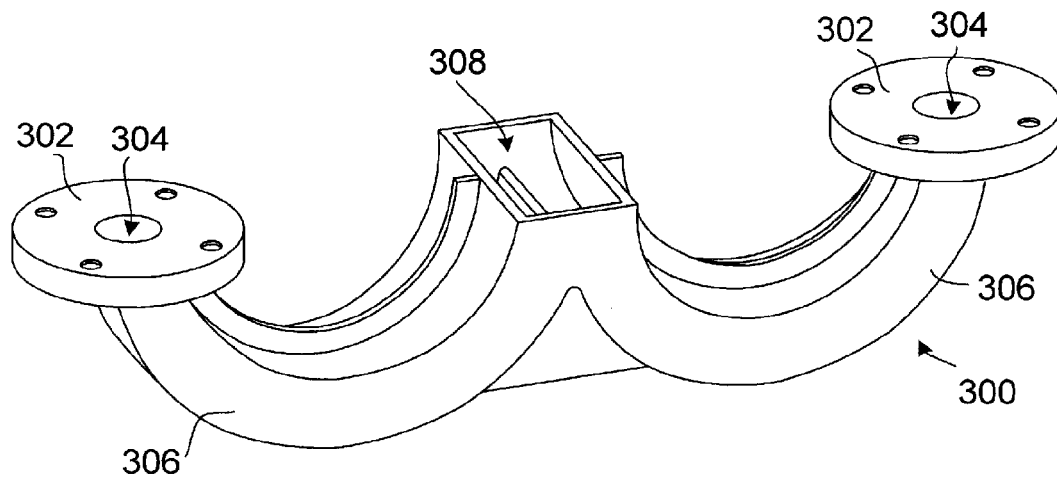


Fig. 8

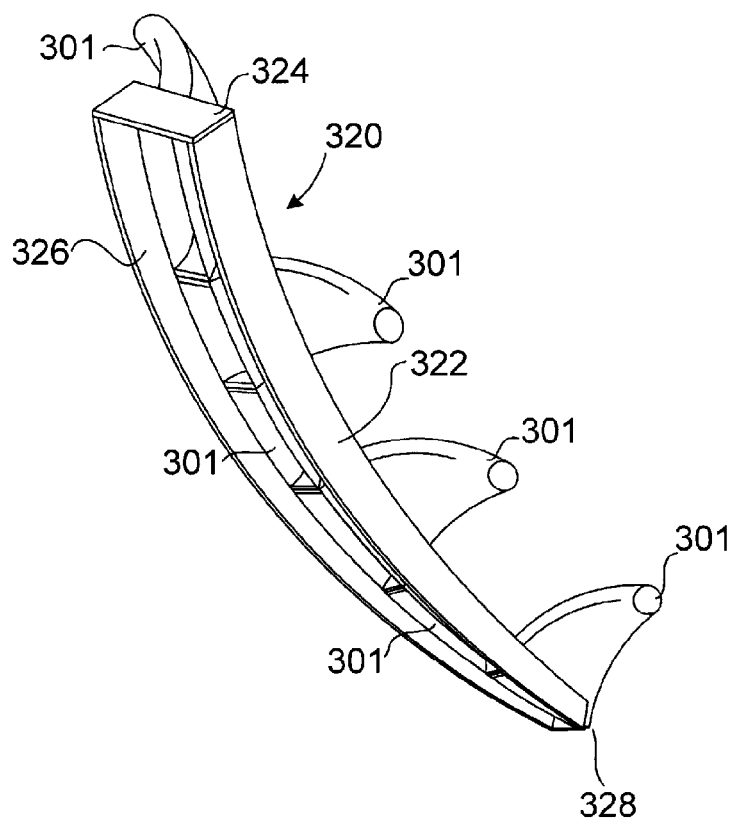


Fig. 9

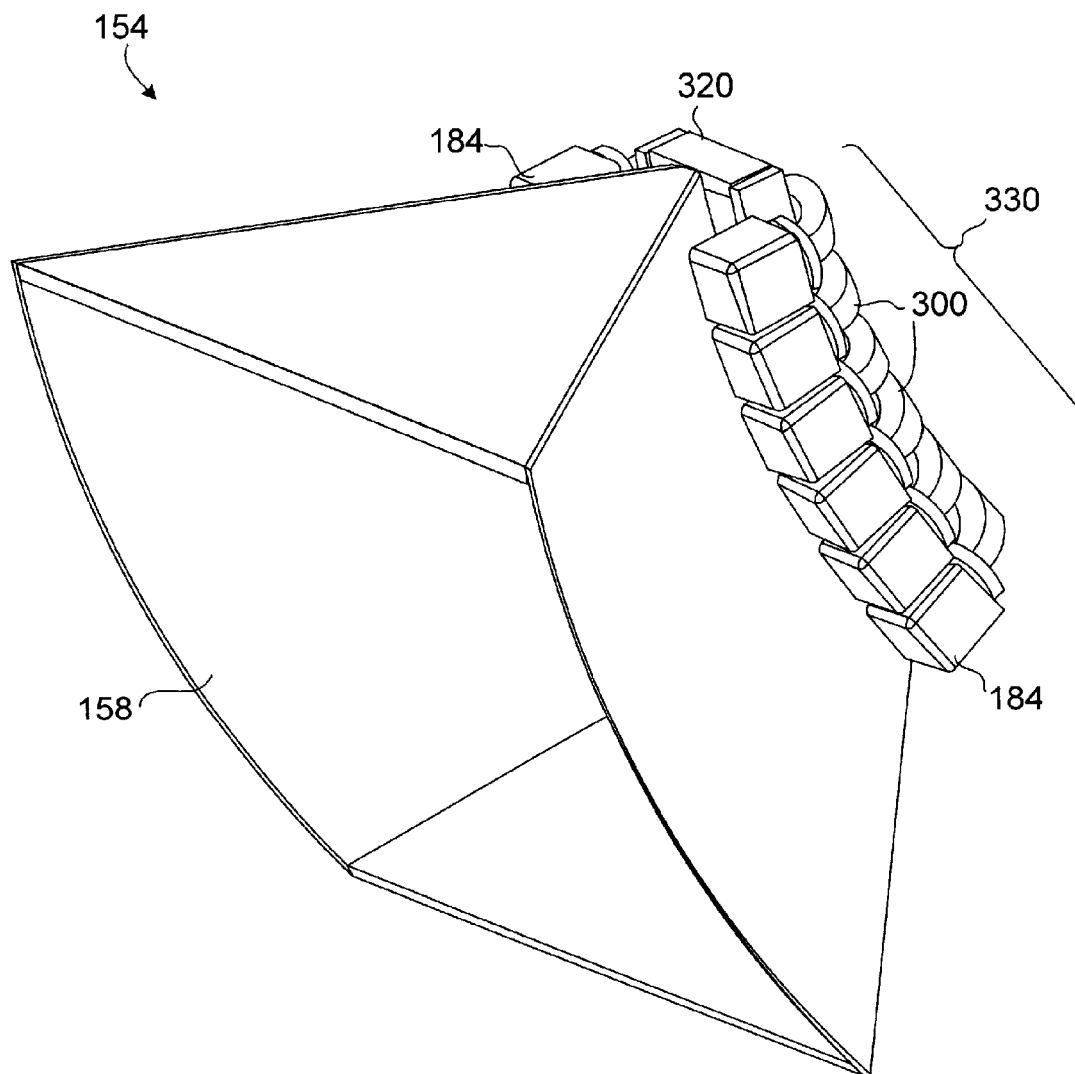


Fig. 10

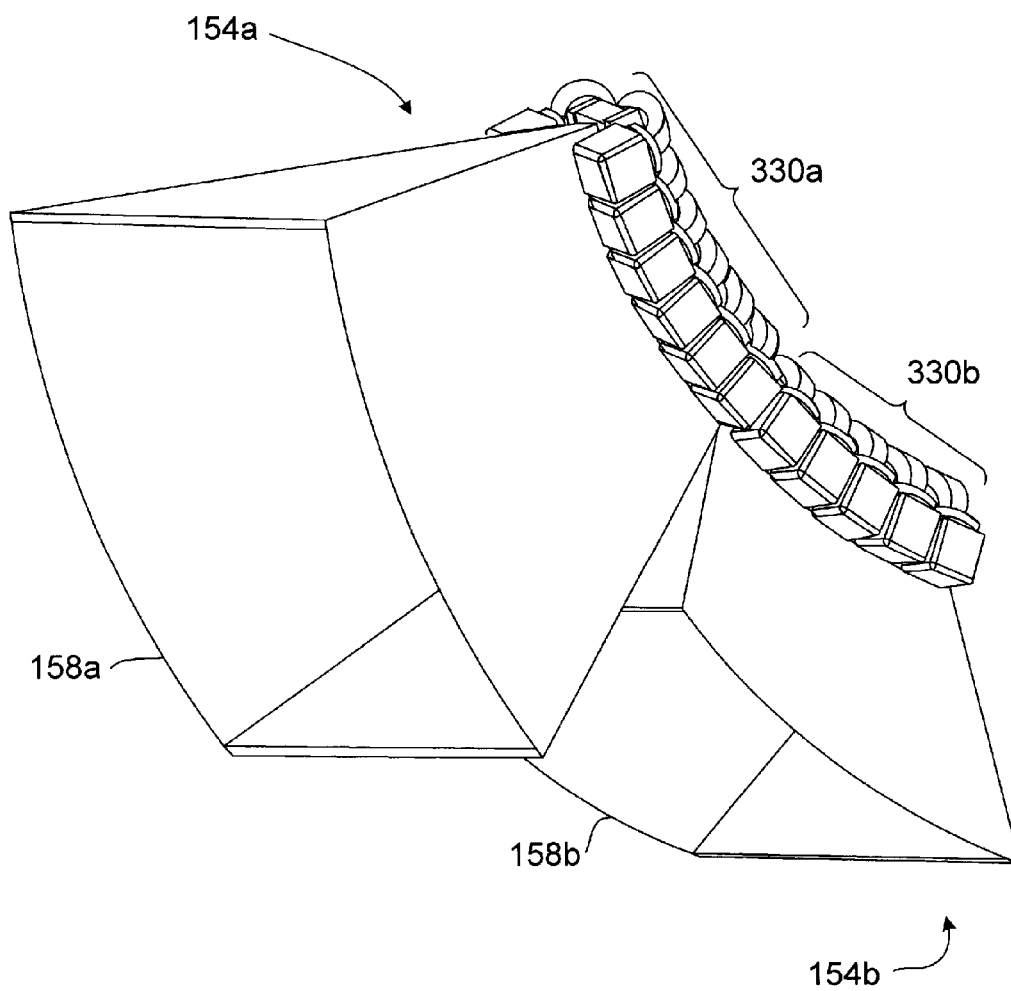


Fig. 11

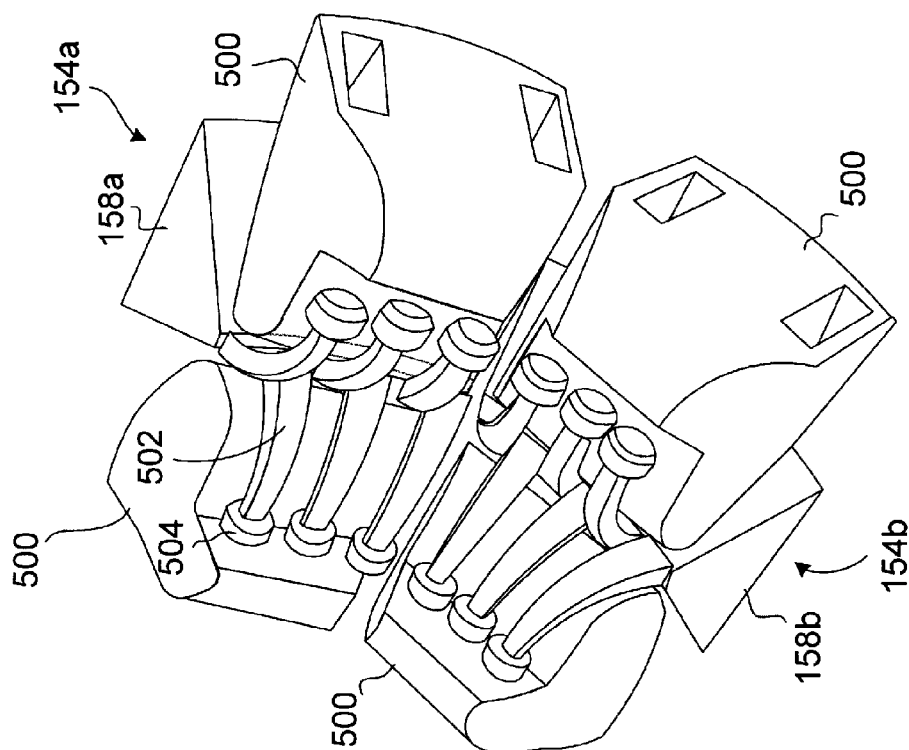


Fig. 12A

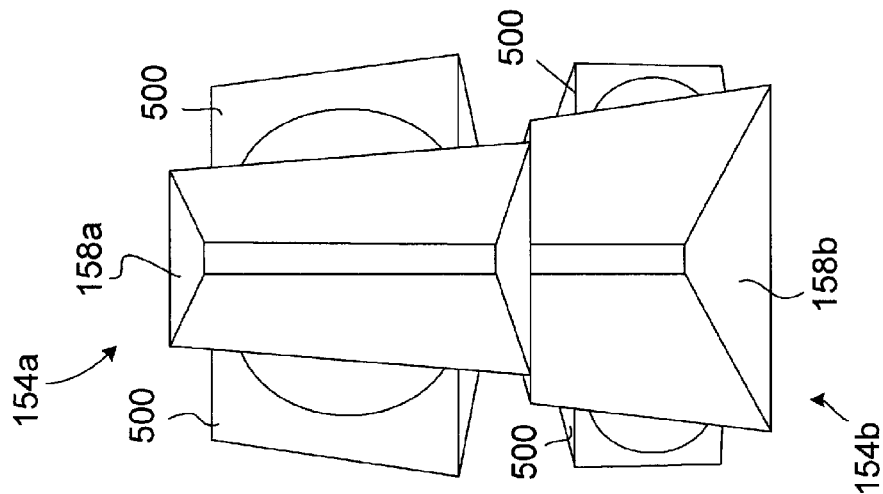


Fig. 12B

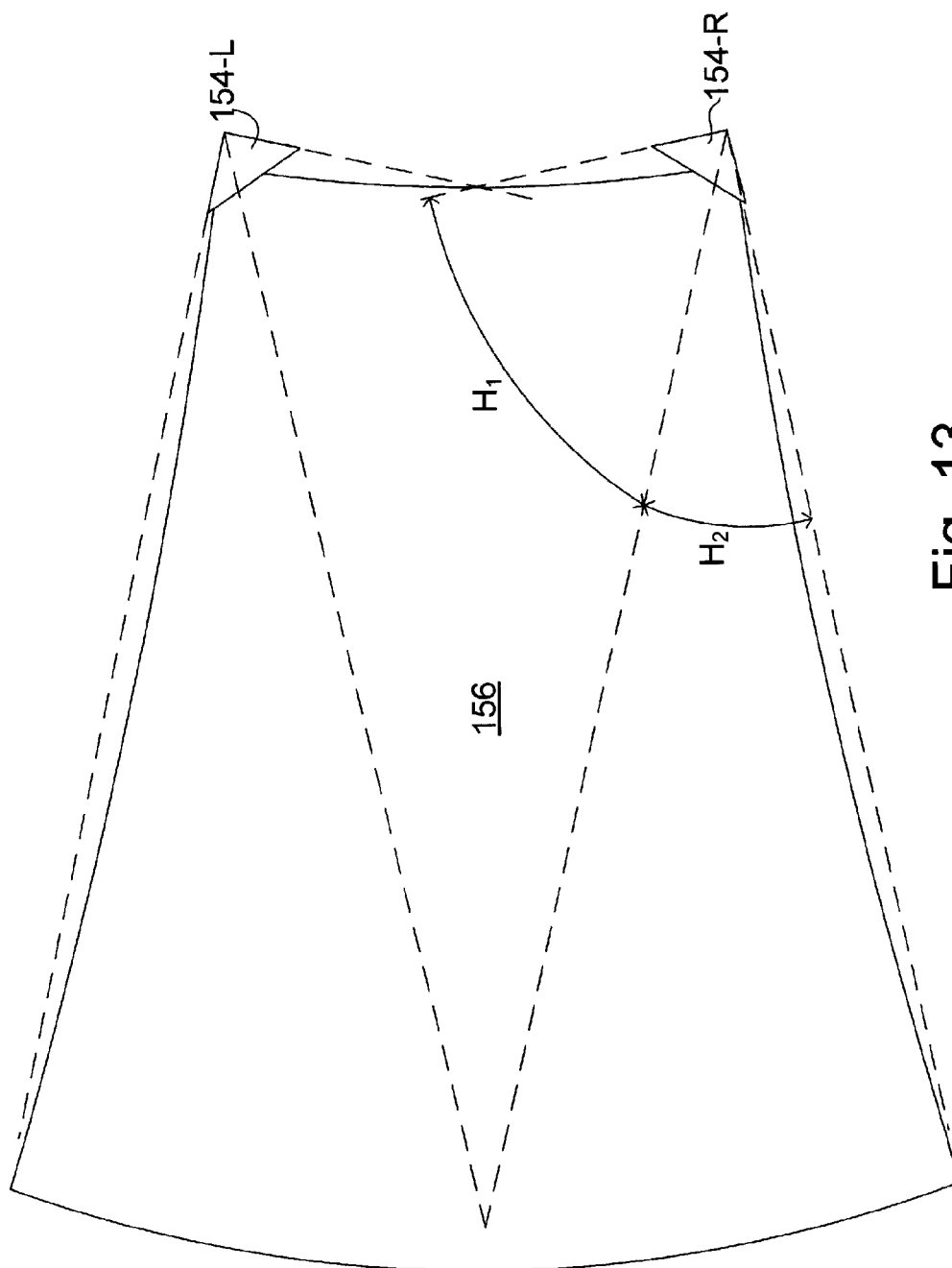


Fig. 13

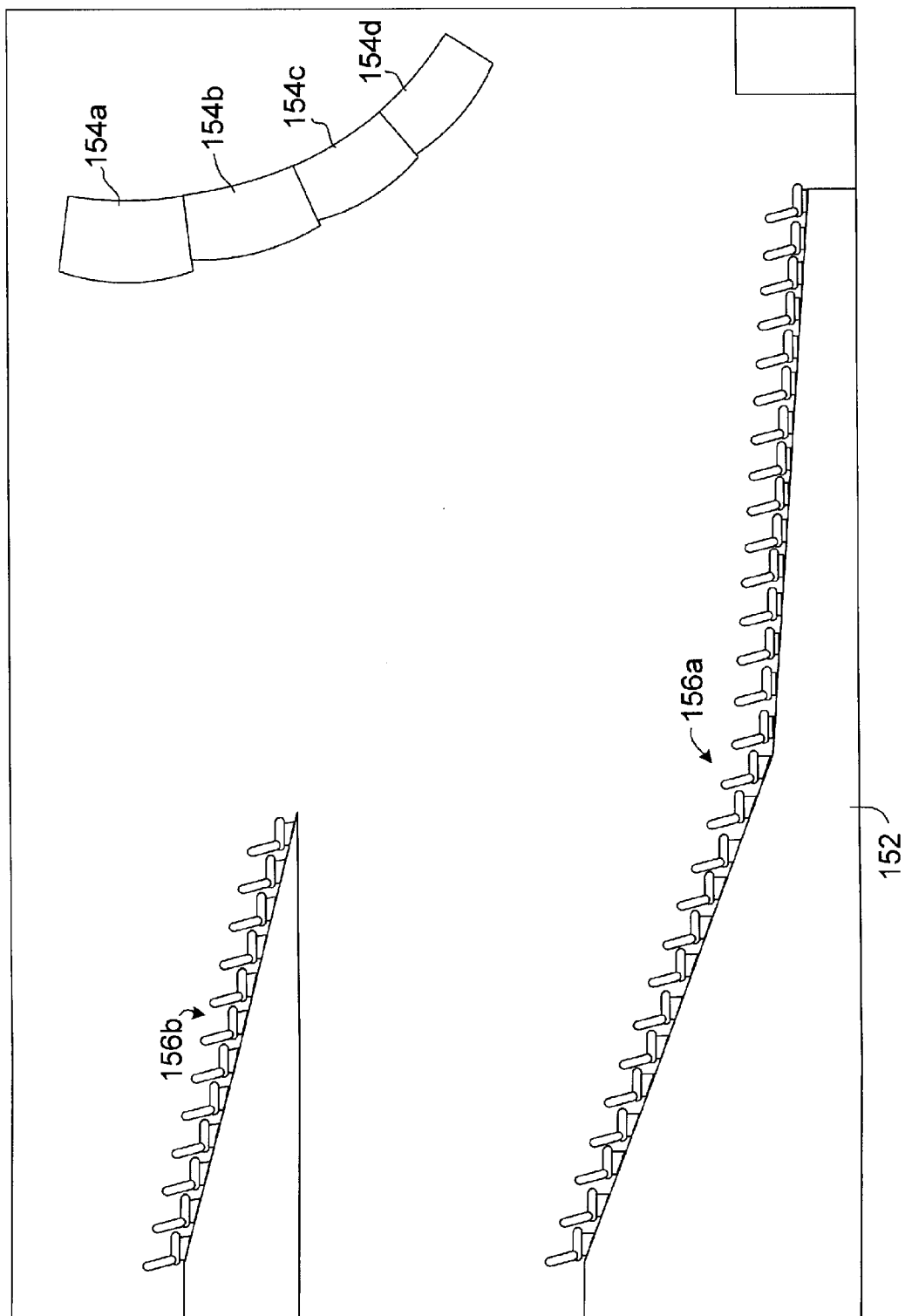


Fig. 14

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AUTOMATED CUSTOMIZATION OF LOUDSPEAKERS

This application is a continuation of U.S. patent application Ser. No. 12/557,885 filed on Sep. 11, 2009.

BACKGROUND

This disclosure relates to automated customization of loudspeakers.

Audio reproduction systems for large venues typically use arrays of modular loudspeakers to produce the level and distribution of sound energy necessary to fill the venue with sound. In some examples, one-off custom loudspeakers that attempt to fill a portion of the venue with a single loudspeaker may be fabricated based on a designer's estimate of the proper dimensions for such a loudspeaker.

SUMMARY

In general, in some aspects, a loudspeaker includes a horn including a first end panel, a second end panel, a first side panel, and a second side panel. Edges of at least the first and second side panels define a diffraction slot opening. A plurality of manifold components each has an output opening coupled to the diffraction slot opening. The output openings of the plurality of manifold components together constitute a diffraction slot source at the diffraction slot opening. A plurality of electro-acoustic transducers are each coupled to an input opening of one of the manifold components.

Implementations may include one or more of the following features. The first and second end panels each include a rigid sheet material. The diffraction slot opening has a convex curvature relative to the inside of the horn. The side panels are curved in a direction corresponding to the diffraction slot opening while remaining straight in a direction perpendicular to the slot opening. The first and second side panels have dimensions when flat corresponding to a section of a planar annulus defined by an inner radius, an outer radius, and an angle. The first and second side panels each vary in length and curvature along the extent of the slot opening. The manifold components each have two input openings and include two acoustic passages, each of the acoustic passages having a first end at a different one of the two input openings and a second end at the output opening, and the acoustic passages each curving away from the output opening in a different direction, such that the input openings are located near opposite sides of the horn. The manifold components each have one input opening and include one acoustic passage having a first end at the input opening and a second end at the output opening, the acoustic passage of each manifold component curving away from the output opening in a direction opposite that of the neighboring manifold components' acoustic passages, such that the input opening is located near the opposite side of the horn from the neighboring manifold components' input openings.

In general, in some aspects, a loudspeaker array includes a first loudspeaker module having a first horn with a first end panel, a second end panel, a first side panel, and a second side panel. Edges of at least the first and second side panels define a first diffraction slot opening curved along a first radius. A first plurality of manifold components each has an output opening coupled to the first diffraction slot opening. The output openings of the plurality of manifold components together constitute a diffraction slot source at the first diffraction slot opening. A first plurality of electro-acoustic transducers are each coupled to an input opening of one of the

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manifold components of the first plurality. A second loudspeaker module includes a second horn having a third end panel, a fourth end panel, a third side panel, and a fourth side panel. Edges of at least the third and fourth side panels define a second diffraction slot opening curved along a second radius. A second plurality of manifold components each have an output opening coupled to the second diffraction slot opening. The output openings of the plurality of manifold components together constitute a diffraction slot source at the second diffraction slot opening. A second plurality of electro-acoustic transducers are each coupled to an input opening of one of the manifold components of the second plurality. The second end panel is adjacent to the third end panel, and the first diffraction slot opening and second diffraction slot opening together effectively form a continuous diffraction slot opening.

In general, in some aspects, a loudspeaker array includes loudspeaker modules, each loudspeaker module having a horn with a horizontal coverage angle and a vertical coverage angle. The horn includes a top end panel, a bottom end panel, a first side panel, and a second side panel. Edges of at least the first and second side panels define a diffraction slot opening curved along a radius. A plurality of manifold components each has an output opening coupled to the diffraction slot opening. The output openings of the plurality of manifold components together constitute a diffraction slot source at the diffraction slot opening. A plurality of electro-acoustic transducers are each coupled to an input opening of one of the manifold components. The array includes a first loudspeaker module having a first horizontal coverage angle, a first vertical coverage angle, and a first radius. A second loudspeaker module has a second horizontal coverage angle greater than the first horizontal coverage angle, a second vertical coverage angle greater than the first vertical coverage angle, and a second radius less than the first radius. The bottom end panel of the first loudspeaker module is adjacent to the top end panel of the second loudspeaker module. The first diffraction slot opening and second diffraction slot opening together effectively form a continuous diffraction slot opening.

Implementations may include one or more of the following features. A third loudspeaker module having the first horizontal coverage angle, the first vertical coverage angle, and the first radius. A fourth loudspeaker module having a third horizontal coverage angle greater than the first and second horizontal coverage angles, a third vertical coverage angle greater than the first and second vertical coverage angles, and a third radius less than the first and second radii. The bottom end panel of the third loudspeaker module is adjacent to the top end panel of the first loudspeaker module. The top end panel of the fourth loudspeaker module is adjacent to the bottom end panel of the second loudspeaker module. The first, second, third, and fourth diffraction slot openings together effectively form a continuous diffraction slot opening.

In general, in some aspects, a method includes obtaining loudspeaker array data identifying a number of loudspeaker modules to be used in a loudspeaker array, horizontal and vertical coverage dimensions for a listening area to be acoustically illuminated by the loudspeaker array, and a distance and angle from the loudspeaker array to the listening area. A division of the listening area into a number of sections corresponding to the number of loudspeaker modules is computed. A horizontal and vertical coverage angle for each loudspeaker module to illuminate a corresponding one of the divisions of the listening area based on the distance and angle from the loudspeaker array to the listening area is computed. A list of the computed horizontal and vertical coverage angles for the loudspeaker modules is outputted.

Implementations may include one or more of the following features. Computing the horizontal and vertical coverage angle for each loudspeaker module includes computing an ideal horizontal and vertical coverage angle, and selecting a horizontal and vertical coverage angle pair from a pre-determined set of coverage angle pairs most closely matching the ideal horizontal and vertical coverage angles. For each loudspeaker module after the first loudspeaker module, computing the ideal vertical coverage angle includes adjusting the computation of the ideal vertical coverage angle to account for any discrepancy between the ideal vertical coverage angle computed for the previous loudspeaker module and the pre-determined vertical coverage angle selected for that previous loudspeaker module.

Advantages include providing a single integrated system that provides a specified acoustic coverage for a particular venue, including at least partially-automated design and fabrication of custom loudspeaker components.

Other features and advantages will be apparent from the description and the claims.

BRIEF DESCRIPTION OF THE DRAWINGS

FIG. 1 shows a process for designing a custom loudspeaker.

FIG. 2 shows a perspective view of a model of a loudspeaker in a venue.

FIGS. 3A, 3B, and 7A show schematic side elevations of a model of a loudspeaker in a venue.

FIG. 4A shows a perspective view of a loudspeaker horn constructed from flat panels.

FIG. 4B shows a side elevation view of the loudspeaker horn of FIG. 4A.

FIG. 4C shows a front elevation view of the loudspeaker horn of FIG. 4A.

FIGS. 5A, 5B, and 6A show flat panels used in constructing a loudspeaker horn.

FIGS. 6B and 6C show a conic section used in defining the flat panel of FIG. 6A.

FIG. 7B shows a schematic plan view of a model of a loudspeaker in a venue.

FIG. 7C shows a side elevation of an intermediate model of a loudspeaker horn.

FIG. 7D shows a front elevation of the intermediate model of FIG. 7C.

FIG. 7E shows a plan view of the intermediate model of FIG. 7C.

FIG. 7F shows a side elevation of a model of a loudspeaker horn.

FIG. 7G shows a front elevation of the model of FIG. 7F.

FIG. 7H shows a plan view of the model of FIG. 7F.

FIG. 7I shows a perspective view of the model of FIG. 7F.

FIG. 8 shows an example transducer manifold element.

FIG. 9 shows a support assembly for components of a loudspeaker with example transducer manifold elements.

FIGS. 10 and 11 show assembled loudspeakers.

FIG. 12A shows a rear perspective view of a loudspeaker assembly.

FIG. 12B shows a front perspective view of the loudspeaker assembly of FIG. 12A.

FIG. 13 shows a plan view of two loudspeakers in a venue.

FIG. 14 shows a side elevation view of a loudspeaker array in a venue.

DESCRIPTION

An automated loudspeaker design system uses a process such as that shown in FIG. 1. In using the system, a user

begins with an architectural model of the venue, and enters design goals, such as the speaker locations, the listening area, the desired bandwidth of the sound, and the sound level to be achieved. A simplified representation of a venue is shown in FIG. 2. The system uses the input parameters to design a custom loudspeaker that will meet the specified design goals. In some examples, the system goes on to supplement the custom loudspeaker with additional components, such as bass loudspeakers, amplifiers, and controllers. The system may also determine configuration parameters for those components, such as equalization curves, connection topologies, time delays, and output limits.

In the automated loudspeaker design process 100 of FIG. 1, the user begins by inputting (102) an architectural model of the venue, represented as a simple rectangular room 152 in FIG. 2. This may be done by loading a file output from architectural design or CAD software or by using a user interface provided by the system to create a model of the venue. One such system is described in U.S. patent application Ser. No. 10/964,421, filed Oct. 13, 2004, and incorporated here by reference. Next, the user specifies (104) where one or more loudspeaker 154 will be located within the venue. By "loudspeaker," we refer to the complete assembly of active electro-acoustic transducers and passive components such as waveguides, ports, and horns that work together to deliver sound to the venue. In FIG. 2, one loudspeaker 154 is shown as a generally trapezoidal box corresponding to the horn, the most visible part of the loudspeaker. Generally, there is a limited choice of where to locate the loudspeakers, due to architecture, aesthetics, or other considerations. In some examples, the automated portions of the system may specify one or more locations for the loudspeakers. The user also specifies (106, 108) the listening area 156 and sound pressure level (SPL) required of the sound system over the listening area. In some examples, the user specifies (110) the topology of a loudspeaker, e.g., whether it will be an exponential horn, a line array with an arc, progressive (spiral), or J-shaped curve, or some other topology. The system described is particularly well-suited to loudspeakers having elongated diffraction slot sources with a continuously curved shape, i.e., a circular curvature or a progressive curve, coupled to a large horn or waveguide. Selection of the loudspeaker topology may also be automated, with the system selecting the best topology to provide the required coverage. Other user inputs (not shown) may include the bandwidth or type of sound to be produced (e.g., music or voice), identification of multiple zones of coverage, or types of audio signal input sources, to name a few examples.

Given these inputs, the system begins to design (112) the first (if more than one) loudspeaker, which consists of a source 160 and a horn 158. As a first step, the system determines (114) initial approximations for the horizontal coverage angle H and vertical coverage angle V over which the loudspeaker will need to project sound to cover the listening area 156. H and V are determined geometrically from the dimensions of the listening area and the location of the loudspeaker. In FIG. 2, due to the unequal length and width of the diffraction slot source 160, V is measured between the top and bottom walls of the horn, based on a projection of those surfaces to their intersection at a hypothetical point source 155 located behind the actual loudspeaker 154, while H is measured from the intersection of projections of the side walls, just behind the back edge of the horn. The vertical coverage angle V is the angle between planes connecting the vertical point source location to the front and back rows of the listening area. The horizontal coverage angle H is the angle between planes connecting the diffraction slot source to the

right and left sides of the seating area. As described below, the values ultimately used may vary from the initial angles suggested by the structure of the venue. For a rectangular seating area, the angle H may vary over the height of the horn, as it will be narrower at the top than at the bottom to cover the same width at the greater distance of the back row. For a seating area that widens with distance (with appropriate dimensions), the angle H may be constant over the height of the horn, the increased coverage with distance matching the increased width of the back of the seating area. Generally speaking, the angle H varies along the height of the horn according to a function based on the variation in coverage area with distance. With a sufficiently large number of drivers, extremely complicated seating areas may be covered quite efficiently.

The system predicts (116) the total radiated power that will be required to provide the specified sound level over the listening area. The power and the coverage angles may be determined independently, or they may be co-dependent. In cases where the diffraction slot source is driven by multiple acoustic transducers, which we refer to as drivers, the user optionally selects (118) a number N of drivers to be used. The number of drivers may be traded off against the wall length L of the horn (see FIG. 4A) to achieve the required radiated power. A horn with longer walls (for given coverage angles H and V) will have a wider mouth, providing control of directivity down to lower frequencies, thus requiring less power to cover a given area than would a shorter horn. A given driver has a fixed amount of power available, so the number of drivers needed is affected by the amount of control the horn provides over power distribution and the impedance matching of the driver to free air provided by the horn. The total length of the source, determined by the number of drivers and their dimensions, also affects the efficiency with which it delivers the power provided by the drivers. In many cases, the dimensions of the horn are constrained by the architecture and by visual considerations, and the number of drivers is dictated by the power required to provide the specified SPL over the listening area. Different models of driver may also be specified to deliver different power levels. Once H, V, N, and L have been determined or input, the system has the information needed to design a source and horn that will provide the specified coverage.

Geometrically, the transducing mechanism may be distant from the aperture or surface that radiates sound, as described below. We refer to the radiating apertures or surfaces that are arranged in a line to form the diffraction slot source as "sources" to differentiate them from the "drivers" that actually transduce electric signals into acoustic pressure. It is the arrangement of these "sources" that is discussed here in reference to FIGS. 1 through 3. Generic speaker symbols are used to represent the sources in FIG. 3 without implying any particular transducer technology or geometric relationship between the drivers and the sources. The precise positioning of the drivers may be independent of the positioning of the sources, and is described later with reference to FIGS. 8 through 10. For this discussion of the algorithmic design of the loudspeaker, the sources may be considered as drivers located directly at the diffraction slot, as illustrated.

For a circular arc-shaped diffraction slot, the N sources are simply positioned (120) in an arc, with the angular extent of the arc corresponding to the total vertical coverage angle V, and the radius dictated by the number of sources N that must be fit into the arc at that angle and their heights.

For a diffraction slot with a progressive curve shape, where the radius of curvature varies with position, there are several methods that may be used to determine the angles and positions of the constituent sources. In some examples, the angles

are determined following an arithmetic or other mathematical progression, such as a geometric series, and their positions are set to form a continuous source along that curve.

In another example, the system may use a model illustrated by FIG. 3A and described below to design the curved diffraction slot and position the sources. The illustration of horn positioning in FIGS. 7A and 7B may also be informative for this discussion. Given H, V, and the required power over the listening area, the system divides (120a) the listening area 156 into N equal-power strips 162i (in the case of FIG. 3A, N=7, so there are seven strips 162a-162g) along the length of the listening area, each spanning its full width, and lays out an array 170 of N sources (160a-160g in FIG. 3A) corresponding to the strips 162i. For a given power available from the drivers and a properly dimensioned horn, each source can illuminate an area, perpendicular to the source's axis, of a particular size (an equal-sized area at a greater distance calls for a narrower horn, constraining the power to a narrower solid angle). The strips 162i represent portions of the listening area that are provided the same sound levels from sources driven by drivers of uniform power. Whether the strips increase, decrease (as shown), or maintain constant depths with increasing distance from the loudspeaker depends on a number of factors, including whether the listening area increases or decreases in width, the angles between the loudspeaker and each strip, and how well the loudspeaker controls dispersion at increasing distances. The sources 160i are aimed (120b) to each cover their respective strip 162i, and arranged to form a continuous curve.

In some examples, the listening area is divided into the equal-power strips as follows. FIG. 3B shows a detail of one strip 162g from FIG. 3A, with additional measurements indicated. When the loudspeaker is projecting to the listening area strip 162g at an angle, the listening area is covered by a projected area 180g that is perpendicular to the direction of radiation, indicated by axis 164g. The projection to the listening area strip 162g is bounded by front and rear planes 182a and 182b. For a small section of the listening area at a downward angle v from the loudspeaker, which we call the coverage area Ac, the projected area Ap(v) is computed from the coverage area Ac(v) and the incident angle θ :

$$A_p(v) = A_c(v) \times \cos(\theta) \quad (1)$$

The power required to provide the specified direct-field intensity Id is equal to the intensity times the area covered. For a loudspeaker that is close enough to the listening area that the angle of incidence varies appreciably from the front of the listening area to the back, the actual coverage areas vary with angle (which in turn varies with position), even while projected areas remain constant with distance, so we compute an integral over the listening area (i.e., between vertical angles v0 at the front and vn at the rear) to find the total required power:

$$W_{req} = \int_{v_0}^{v_n} I_d(v) A_c(v) \cos(\theta) dv \quad (2)$$

The function defining $A_c(v)$ in the integral (2) depends on the geometry of the venue and to the loudspeaker location, as does θ . Direct-field intensity I_d may also be a function of the position in the listening area, or it may be constant. The particular solution to the integral will therefore be based on the inputs to the process 100, and it may be solved numerically or algebraically, depending on the capabilities of the system carrying out the process 100.

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The required power W_{req} is divided by the nominal power W_s provided by one driver to find a lower bound for N, that is:

$$N \geq \frac{W_{req}}{W_s}. \quad (3)$$

The user may specify a larger value of N, or may choose to at least start with the computed minimum value. The listening area is then divided up into the equal power strips **162i** by solving the integral (2) for values of v_1 and v_2 that define an area covered by the power W_s of one driver:

$$W_s = \int_{v_1}^{v_2} I_d(v) A_c(v) \cos(\theta) dv. \quad (4)$$

Starting at the rear of the listening area, where v_2 is known (i.e., for the first strip, $v_2 = v_n$), the integral (4) is solved iteratively for each strip **162i**, with the previous slice's lower angle v_1 becoming the next slice's upper angle v_2 . If the user has specified a larger N than that required by (3), then W_s in (4) can be replaced by W_{req}/N_{user} . As the angles vary, the areas of the actual coverage areas will vary, while the projected coverage area is the same for each actual coverage area. In some examples, such as where the loudspeaker is low to the ground and the angles of incidence quite severe, additional numeric calculations may be needed for each source and coverage area, rather than the simplifications just noted. In the case of a single horn, the actual coverage areas are rectangular (or nearly-rectangular) strips spanning the width of the listening area and having varying depths, as shown in FIG. 3. The dimensions used in FIG. 3 are arbitrary, and selected only to illustrate the process. If multiple loudspeakers are to be used, the strips may cover less than the full width of the listening area. Once the strips have been found, the sources are each aimed (**120b**) to cover their corresponding strip.

An iterative process may also be used to determine the positions of the sources and define the curve of the total diffraction slot source. A first source **160a** is positioned with its top edge defining the top end of the curve. This source begins the curve just above the plane connecting the point source **155** to the back row, so that the drop off in energy (typically -6 dB) at the edge of the projected coverage area falls outside the listening area. The source **160a** is tilted such that its axis **164a** passes through the center of the rear-most strip **162a** defined in (4), thus defining the source's pitch angle. The pitch angle for each source will correspond to the angle v at the center of the corresponding strip, used in the calculations above. From the top edge position, pitch angle, and the dimensions of the source, the location of the first source's lower edge is known. The next source is positioned with its top edge at the lower edge of the previous source, and the above steps are repeated. This continues until all sources are positioned, producing the progressive curve seen in FIG. 3. The final source will be located with its bottom edge extending just beyond the plane connecting the point source **155** to the front row. The drop-off in power at the edges of the coverage areas is compensated for between sources by the overlap that results in the coverage of adjacent sources. The areas computed above will result in angles between sources that properly overlap the projected areas of each source, using the overlapping coverage of the sources to provide even coverage in the transition regions. The same process could be applied starting with the lowest source **160g** and the front-most strip **162g**, and the process may be varied to meet other goals, if, for example, uniform intensity over the listening area is not desired.

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Once the diffraction slot source provided by the arc of sources is defined, the horn walls are designed to confine the sound radiated by the sources to the listening area. A generalized horn **158** is shown in FIGS. 4A-4C. Independently of the source curvature, the horn **158** may have a complex shape that is narrower and deeper at the top, for projecting sound to the back of a venue, and wider and shallower at the bottom, for spreading sound over the front of the venue. This allows a single loudspeaker to cover a significant portion of the listening area. In other examples, the horn maintains the same shape from top to bottom. To define the walls of the horn **158**, flat top and bottom panels **166**, **168** are positioned (**122**) in planes separated by the vertical coverage angle V to align with the top and bottom of the listening area **156** (FIG. 2) (more generally, the panels **166** and **168** may be referred to as end panels, as the loudspeakers may be installed in various orientations). Side walls **200** of the horn **158** are next defined (**124**) to align with the sides of the listening area **156**. The side walls **200** are separated by the horizontal coverage angle H , which as noted above, may vary with height. To accommodate the curved source defined in the previous steps, the side walls **200** are also curved. In the horn fabrication process described below, the side walls are formed from flat panels and bent to the needed curved shape. In some examples, the ideal curve of the side walls puts too much stress in the material, so the curve of the side walls is softened (**124a**) from the ideal shape as needed. If additional loudspeakers are called for, they are designed (**113**) following the same process, iteratively or in parallel with the first loudspeaker.

Once the horn geometry has been determined, it may be refined. Sound system simulation software may be used (**126**) to determine equalization parameters and predict the radiation pattern of the loudspeaker. The predicted radiation pattern is then offered (**128**) to the user, either graphically, as described, for example, in U.S. patent application Ser. No. 10/964,421 or through an audio simulation, as described, for example, in U.S. Pat. No. 5,812,676, both incorporated here by reference. The user chooses (**130**) to accept the design to or make changes to one or more of the parameters (H , V , N , or L , or one or more of the original input parameters). For example, changing the number of drivers, the wall length, or the coverage angles may result in a smoother radiation pattern. If too many drivers are specified for the budget, the user might decrease the specified sound level; alternatively, if there is room in the budget, the user might directly specify more drivers or might raise the specified sound level. If changes are made, the process iterates the source and horn design steps to produce and simulate a new design. In some examples, the iteration is also automated. If the predicted radiation pattern does not match the input parameters within some threshold, the process uses the difference to refine the values of H , V , N , or L , and then repeats the appropriate steps to alter the shape of the source and horn according to the refined values. This may repeat until the values converge, that is, the deviation from the target, improvement per iteration, or the changes called for to further refine the design are less than some threshold. Once the horn shape is accepted (**130**), the side walls are mathematically (analytically or numerically) transformed (**132**) into flat shapes. Plans for the top, bottom, and side walls are output (**134**) for fabrication (**136**), described below.

In some examples, the loudspeaker design process **100** is paired with a system configuration process **138** that uses the same or similar inputs and outputs configuration parameters such as equalization settings, time delays, limiter levels, bills of material, schematics, and the like. An automated process for system configuration is described in U.S. Pat. No. 7,206,

415, incorporated here by reference. Configuration 138 may also be performed manually. Finally, the system is installed (140) and put to use.

A particular loudspeaker well-suited for customization and, in particular, the automated design process described above, is shown in FIGS. 5 through 10. FIGS. 5A, 5B, and 6A show flat panels that might be designed according to the process described above and assembled to form the horn shown in FIGS. 4A-4C. In particular, the side panel 200 in FIG. 6A is joined to top and bottom panels 166 and 168 like those in FIGS. 5A and 5B to form a horn for use with a circular arc source. The process of determining the shape of these panels is described below. Creation of a horn for a progressive curve from similar materials is described farther below with reference to FIGS. 7A through 7I. A set of modular manifold sections, such as those shown in FIG. 8, couple to the transducers' respective radiating apertures or surfaces and are assembled on a rigid, curved keel shown in FIG. 9 to form the curved diffraction slot source that is joined to the horn 158 to form the complete loudspeaker 154, shown in FIG. 10. While loudspeakers of this type may be designed and built manually, they are well-suited to the automated process described herein, which deterministically designs the speakers based on the input performance and venue parameters.

The horn 158 is formed from four surfaces that begin as flat panels and are joined to form the complex curvature needed. As shown in FIGS. 5A and 5B, the top and bottom panels 166 and 168 are triangles (with their tips truncated) with side edges 202, 204, 206, and 208. The central angles of the triangles determine the horizontal coverage angle H of the horn. If the triangular end panels 166 and 168 are identical (not as shown), then the horizontal angle will be constant along the height of the horn. If the end panels are not identical, the curvature of the side walls will be more complex and H will vary along the height of the horn. One end may be narrower and deeper, while the other end is wider and shallower, though width and depth may be independent. Such a complex curvature provides the long and narrow top and short and wide bottom described above. In an even more complex curvature, the paired edges 202 and 204 of the top panel 166 may be different lengths from the edges 206 and 208 of the bottom panel 168, causing the wall length L to vary along the height of the horn. This may provide, for example, a horn with a narrow end and a wide end, but with the same depth at both ends. In the example shown, the triangles and therefore the horn are symmetric about one axis, but this is also not a requirement. The horn may be asymmetric in the horizontal plan, in which case the opposite edges 202/204 and 206/208 of each triangle will be different.

FIG. 6A shows the flat side panel 200 that will be joined to the triangular end panels 166 and 168 to form a horn fitting a source with a circular curvature. The panel 200 is a section of an annulus, defined by an inner circumferential section 210 with a radius R_1 from a center point 211, an angle V_f between two edges 212 and 214, each of length L (the depth of the horn), and an outer circumference 216 at a radius R_2 . Joining two panels 200 such that the inner edges 210 of the two panels remain parallel while the panels diverge at the horizontal coverage angle H imposes a curvature on the panels along their vertical extent. As shown in FIG. 6B, the side panels 200 for such a horn each correspond to a section of the surface of a section of a cone 220 (only part of the cone is shown). FIG. 6C is the same as FIG. 6B with only one panel 200 shown for clarity. The cone has an apex 221, and the surface defining side wall panel 200 is a band with its edges 212 and 214 lying on lines 222 and 224, and with the inner circumferential edge

210 and outer circumferential edge 216 forming parallel circular arcs on the surface of the cone 220.

The dimensions of the side panel 200 are determined by certain aspects of the cone 220, which are in turn determined by the starting values H, V, N, and L, defined earlier. In FIG. 6B, a horizontal plane 228 defines the inner edge 210 by its intersection with the cone. The plane 228 contains the arc of the curved diffraction slot source (not shown), which will normally be vertical when installed. The side panel 200 extends away from the plane 228 towards the base of the cone. In FIG. 6B, the second side panel is offset a small amount from the cone, representing the width of the diffraction slot source—a second cone corresponding to the second panel is not shown.

The angle between the side panel 200 and the plane 228 is $H/2$, as each side panel 200 is one-half the horizontal coverage angle H from the center plane 228. Because plane 228 is perpendicular to the axis 230 of the cone, $H/2$ is also the base angle of the cone. The vertical coverage angle V of the horn matches the angle in the plane 228 between planes defining the end panels 166 and 168, which intersect at the axis 230 (while the planes aren't shown, they correspond to the near and far faces of the illustrated section of the cone 220). As noted, the length of the edges 212 and 214 is L, directly defining the length of the horn walls. (The far edge 214 of the lower panel 200 is obscured, but both edges 212 and 214 are visible and labeled on the upper panel and in FIG. 6A.)

The inner circumferential edge 210 connects a number of line segments 210a matching the number of drivers N, each of which has a length B. Projecting a perpendicular segment 234 from the center of one segment 210a back to the axis 230 of the cone forms a right triangle having a base of length $B/2$ and a hypotenuse 238. The length R_s of the hypotenuse 238 is the radius in the plane 228 of the inner circumferential edge 210 when the panel 200 is bent to conform to the cone. The angle of this triangle, at the axis 230, is

$$\frac{1}{2} \frac{V}{N}$$

(as the angle V is divided into N segments, and this triangle further divides one of those in half), and the base length is $B/2$, so the length of hypotenuse 238 is found as:

$$R_s = \frac{B}{2 \sin\left(\frac{1}{2} \frac{V}{N}\right)} \quad (5)$$

The value of the inner radius R_1 and of the angle V_f between the edges 212 and 214 of the flat panel 200 in FIG. 6A are found from the fact that when the conic surface is laid flat, the edge 210 is remains circular, but with a different radius R_1 than that it had in the cone, R_s . As a result, the arc length of edge 210 is the same in both figures, thus:

$$R_1 V_f = R_s V \quad (6)$$

The segment of the line 222 from the corner of the panel 200 to the apex 221 also has length R_1 . This segment can be seen to form a right triangle with the radius 238b and the axis 230 of the cone. Knowing the angle $H/2$ between the plane 228

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and the surface of the cone, the length from the apex **221** to the edge **210** is found as:

$$R_1 = R_s \frac{1}{\cos(H/2)}. \quad (7)$$

Solving (4) and (5) for V_f

$$V_f = V \cos(H/2). \quad (8)$$

Combining (5) and (7), R_1 is also found in terms of the inputs:

$$R_1 = \frac{B}{2 \sin\left(\frac{1}{2} \frac{V}{N}\right) \cos(H/2)}. \quad (9)$$

Given V_f , R_1 , and L , the panel **200** can be cut from a flat sheet, as it is simply a section of an annulus having inner radius R_1 , outer radius $R_2 = R_1 + L$, and included angle V_f .

The horn shown in FIG. 6B maintains a constant angle H , such that the top and bottom panels **166** and **168** are identical. Once a panel is designed according to the cone's geometry, however, it may be used in a horn having a varying horizontal angle by using top and bottom panels with differing angles. To design a horn that varies in length L along its height, the outer radius R_2 may be varied with position, independently of whether H will be varied as well. In such a case, the outer radius R_2 may vary monotonically from one end to the other, that is, it will increase or decrease, depending on point of view. If L and H are varied together, such that the walls lengthen as the horizontal angle decreases, a horn may be formed with a varying angle but still having a rectangular mouth, or even having a mouth that gets wider as the angle gets narrower.

A second method of forming the horn sides begins with a series of flat shapes. This method is useful, for example, with a J-curve, progressive curve, or other complex source. As shown in FIGS. 7A through 7E, the horn **158** can be conceptually divided into a series of stacked sub-horns **158i**, one for each driver. FIGS. 7A and 7B show only the top and bottom sub-horns **158a** and **158g** (for a seven driver horn), for clarity. It should be noted that these figures are not to scale within or between figures. Each sub-horn has its own horizontal and vertical coverage angles, H_i and V_i , based on the relationship between its location and the corresponding strip **162i** of the listening area, explained above with regard to FIG. 3. A stack of such sub-horns **158i** is modeled as shown in FIGS. 7C through 7E. Initially, each sub-horn is modeled as having a rectangular mouth, formed by smaller, flat versions of the side walls and its own triangular top and bottom panels. In a simplified case, the side walls may be initially modeled as trapezoidal or as sections of annuli found as described for a circular arc source, above, each with an N of 1 and H , V , and L set according to the coverage area of strip **162i**. As seen in FIGS. 7A and 7B, the top and bottom panels are positioned to begin at the upper and lower edges of their respective sources **160a**, **160g** (determined at steps **120a** and **120b** of process **100**, above) and align with the top and bottom ends of the corresponding slices of the listening area (as noted with regard to the source locating process, the actual coverage areas are typically extended just beyond the limits of the listening area, to account for the drop-off in power at the edges of the coverage areas). Dashed lines corresponding to these coverage areas are shown for sources **160a** and **160g** in FIG. 7A. The angle between these lines, projected back to

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their intersection, will be the V_i for that sub-horn (the intersection of the nearly-parallel top and bottom lines for strip **162a** would be far off the page, so those lines are shown truncated). It should be noted that because all three axes in the figures are drawn to different scales, as are the different components, the geometrical relationships in the figures may not match a real-world implementation. Joining together such a stack of sub-horns with different vertical coverage angles allows a more uniform frequency response over the entire listening area.

Once the stack of sub-horns is modeled, their trapezoidal side-walls are deformed to meet at their edges and corners to form a sheet, and curved to provide the sheet with a continuous curvature, as shown in FIGS. 7F through 7I. Dashed lines show the edges of the deformed and merged segments. Various techniques may be used to join the panels, such as averaging lengths, matching mid-points, or moving each bottom corner out to the adjacent top corner, to name some examples. The top and bottom panels of each modeled sub-horn (except the outermost top and bottom panels) are removed, leaving a single open mouth. In some examples, one or more of the intermediate top and bottom panels or smaller bracing members may be retained to lend structural stability to the horn and maintain the designed horizontal angle H . If the curvature of the side panel generated by this process exceeds what is possible or structurally sound for the material to be used, then the curvature may be modified, such as by widening or narrowing the horn to remove inflection points (where curvature changes from convex to concave) or decreasing radii. In some examples, it has been found that extremely complex curvatures are possible, as long as there remains at least one direction at any given point where the local relative curvature is zero.

To form a complex curvature from flat panels, the side panels **200** are fabricated from a thin, flexible material having enough stiffness to maintain a relatively rigid finished assembly, once the ends are fixed to the less-flexible top and bottom panels, which are in turn fixed to the keel. In some examples, PVC sheets having a thickness of 3.18 mm (0.125 in) have been found to have a suitable stiffness. Aluminum sheets having a thickness of 1.27 mm (0.050 in) have also been found to have a suitable stiffness, though metals may not have enough internal loss to prevent sound from escaping through the walls. The triangular end panels **166** and **168** are not curved in this example, and can therefore be made from stiffer or thicker flat material. We have found that plastic side walls prevent even very high pressure sound from escaping through the walls, despite the thin material. We believe this may be due to the stressed curvature of the side walls increasing their rigidity once assembled, together with a high internal loss in the material. By "flexible," we refer to material that can be bent to the desired shape while remaining within the elastic limits of the material, that is, there is no plastic deformation or cracking when bending the sheets to shape. In contrast, by "rigid" we refer to material having sufficient stiffness that it will impart the stress needed to maintain the bent shape of the parts it supports without being bent or otherwise deformed itself. It is preferred that all the materials used in the horn be environmentally stable. In some examples, a horn may need a small vertical angle and flat top and bottom ends, resulting in side walls with minimal curvature in the vertical direction. In order to add the stress needed for stiffness in this case, the side panels may be given a curvature in the horizontal direction, i.e., the horn width flares outward along a curve.

In some examples, the top and bottom panels may be curved. In such a case, the side panels may also be curved or may remain flat, depending on the particulars of the horn. For

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example, for very small vertical angles, the diffraction slot may have little curvature, resulting in generally flat side panels, while the top and bottom panels may be curved to flare away from the diffraction slot. In such a shape, the ends of the side panels would be curved, rather than straight, to conform to the curved end panels. If the curved end panels are formed from a rigid material, then the structure may remain as described above. On the other hand, if the end panels are formed from a flexible material, then the side panels may need to be rigid in order to support the curvature of the end panels. Alternatively, the joint between the keel and the curved edges of the side panels may impart sufficient rigidity that all four panels may be flexible and hold each other to shape, in tension.

The audio source for use with the horn **158** in the complete loudspeaker is a curved diffraction slot driven by a set of electro-acoustic transducers. The transducers may be, for example, compression drivers, cone-type transducers, electro-static transducers, or other types of electro-acoustic transducers. The diffraction slot source is formed by stacking a number of modular manifold components that each couplemated transducers to a slot-shaped outlet. An example manifold component **300** is shown in FIG. **8**. In this example, each manifold component **300** has two mounting plates **302** where compression drivers (not shown) are attached. The compression drivers direct sound energy into openings **304** into ducts **306**. The ducts **306** end at a rectangular aperture **308**. When a number of manifold components **300** are stacked, as shown in FIG. **9** (with a different example manifold component **301**) and **10**, the apertures **308** together form the diffraction slot source. By moving the transducers away from the diffraction slot, and to the space behind the side walls of the horn, where there is more room, the manifold components **300** allow larger transducers to be used than if the transducers were directly located at the diffraction slot. While the example of FIG. **8** shows a two-sided manifold, single-sided manifold components may also be used, alternating sides as shown in FIGS. **9** and **12**, to accommodate even larger diameter transducers.

In some examples, a rigid keel **320**, shown in FIG. **9**, is used to provide structure for aligning the manifold components in the desired curvature and joining them to the horn **158**. In the example of FIG. **9**, single sided manifold components **301** are shown, as opposed to the two-sided components **300** of FIG. **8**. Each manifold component **301** couples a single driver (not shown) to the diffraction slot, with alternating manifold sections bending to opposite sides of the loudspeaker, accommodating drivers with a larger diameter. The keel **320** may also provide a structural "back bone" for the horn. In some examples, the keel is assembled from steel panels **322**, **324**, **326**, **328** which provide a rigid support for anchoring the other components and hanging the assembled loudspeaker. In some examples, the keel is formed from two curved parts that simply sandwich the manifold components. In the example of FIG. **9**, the manifold components are not anchored to the sides of the keel—the sides of the keel project forward of the manifold openings and will be positioned to either side of the rear of the horn, while the manifold components will meet the edges **210** of the side walls.

The components described above are assembled to form a complete loudspeaker **154**, as shown in FIG. **10**. The keel **320** is attached to the base of the horn **158**. A stack **330** of manifold components **300** is attached to the keel **320** with their apertures **308** mating to the opening in the horn to form the diffraction slot source. Each manifold component is mated to at least one driver **184**. In the example of FIG. **10**, the two-sided manifold components of FIG. **8** are used, so two drivers

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are used with each manifold component, though only the top one on the far side is visible in the figure.

The area covered by the loudspeaker **154** is determined by the dimensions of the horn **158**. Given a required coverage pattern, the necessary dimensions of the horn **158** can be determined as described above. In some examples, as shown in FIG. **11**, multiple loudspeakers **154a**, **154b** having horns **158a**, **158b** with different horizontal and vertical angles and lengths, and manifold stacks **330a**, **330b** with the same or different numbers of drivers, may be stacked. This allows smaller individual components to be used, simplifying fabrication, shipping, and storage of components. In some examples, the progressive-curve process described above would result in too great a curvature in the side panels, so two or more stacked loudspeakers are used, each providing the coverage of the corresponding portion of the modeled single horn. When loudspeakers are stacked in this way, the top and bottom panels of adjacent horns are sufficiently thin that the adjacent diffraction slots are close enough together that they combine to act as a single continuously-curved diffraction slot.

As noted above, once the dimensions of the horn panels **166**, **168**, and **200** are determined, the system outputs **(134)** plans, i.e., instructions for fabricating **(136)** the panels and, in some examples, the keel, from flat stock. In some examples, the system directly outputs machine-readable instructions for use by an automated fabricator, such as a 2-dimensional CNC milling machine. The panels and keel may be cut on-demand, decreasing the amount of unique inventory that must be maintained for an installer to offer a large, and even unlimited, variety of custom horns. Whether custom-cut or pre-manufactured, the panels may be shipped flat and assembled at the point of installation, decreasing the volume of space required for storage and shipping. Various standard construction methods may be used to join the panels and other components, depending on the materials used, such as screws, adhesives, or press fittings.

In one example, the ends of the side panels are fixed to the top and bottom panels with screws, and additional plates are used to fix the top and bottom panels to the keel. Prior to fixing the top and bottom panels to the keel, the side panels are under decreased stress and may relax to a different shape, such that the edges forming the diffraction slot opening do not remain parallel. To assemble the horn, one end panel, i.e., the top or bottom panel, is fixed to the keel, and the other panel is then drawn toward the other end of the keel. As the free end panel is pulled into position, it forces the side panels to deform to the desired shape. In some examples, the side panel edges come into the desired curvature at the keel and fit into prepared grooves in the keel structure. Once the free end panel is secured to the keel, the bent edges of the side panels remain pressed into the groove in the keel, forming a tight fitting without the use of additional fasteners, though a sealant may be needed to fully seal the joint. In other examples, such as where the keel support structure is wider than the slot opening, the bent edges of the side panels do not meet the keel structure, but couple only to the manifold openings.

The large horns described herein, in combination with arrayed transducers, achieve good directivity control over a wide frequency range, in some examples going as high as 20 kHz and as low as 400 Hz, 300 Hz, or even 250 Hz. For a complete sound reproduction system, modular bass loudspeakers may be added, as shown in FIGS. **12A** and **12B**. In the example of FIGS. **12A** and **12B**, modular bass loudspeakers **500** are shaped to fit partially behind the flare of the horns **158** of the loudspeakers **154**. The bass loudspeakers **500** are shaped to fit behind horns having any flare angle or curvature

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within the range of possible horns, so the same bass loudspeakers **500** are used with any horn made using the above process. For example, in FIGS. **12A** and **12B**, the same bass loudspeakers **500** are used with a relatively narrow upper horn **158a** and a relatively wider lower horn **158b**. In the example of FIG. **12A**, each manifold section **502** couples a single driver **504** to the diffraction slot, alternating sides, as in FIG. **9**.

By providing good reproduction with controlled directivity down to 400 Hz or less from a single type of loudspeaker, and using the bass loudspeakers only for lower frequencies, with some overlap possible, the crossover point of the system is below or in the lower end of the range of human voice (around 300 Hz to 4 kHz). This avoids having a discontinuity in directivity within the middle of the voice range, as may be the case when a crossover must be higher in the voice range, such as around 1-2 kHz. The ability to position the bass loudspeakers close to the horn loudspeakers and in a precisely controlled relative position further enhances the directivity transition at the crossover point and allows good phase and timing alignment over the full angular range.

FIGS. **13** and **14** show two installation examples that combine several of the alternatives mentioned above. In FIG. **13**, two asymmetric loudspeakers **154-L** and **154-R** are used, one on each side of the listening area **156**. The inner side walls (the side walls facing the middle of the listening area) of the loudspeakers have a large horizontal angle H_1 to reach the front-center of the listening areas, while the outer side walls have narrow horizontal angles H_2 to confine the sound to the outer edge of the listening area with sufficient sharpness to reach the rear. The angles H_1 and H_2 are unequal because the diffraction slot is aimed at the rear-center of the listening area **156**, rather than centered between the inner and outer walls of each horn, to provide even sound coverage. As in the other figures, axes are not to scale and the angles and distances in FIG. **13** are exaggerated to show their differences.

In FIG. **14**, a long array of loudspeakers **154a-154d** is used to provide an extremely long diffraction slot for covering a large venue **152** with two listening areas **156a** and **156b**. The uppermost loudspeaker **154a** reaches the back of the balcony **156b**, while the lower three loudspeakers **154b-154d** cover the lower floor **156a**. In such a configuration, circular-arc modules of differing radius may be arrayed to achieve the effect of a long spiral array while using the simpler shapes of the circular-arc horn and keel.

Other implementations are within the scope of the following claims and other claims to which the applicant may be entitled.

What is claimed is:

1. A loudspeaker comprising:
 - a horn comprising a first end panel, a second end panel, a first side panel, and a second side panel, edges of at least the first and second side panels defining a diffraction slot opening;
 - a plurality of manifold components, each comprising at least one acoustic passage and an output opening connected to the diffraction slot opening, the output openings of the plurality of manifold components together constituting a diffraction slot source at the diffraction slot opening; and
 - a plurality of electro-acoustic transducers, each connected to an input opening of one of the manifold components.
2. The loudspeaker of claim 1 wherein the first and second end panels each comprise a rigid sheet material.
3. The loudspeaker of claim 1 wherein the diffraction slot opening has a convex curvature relative to the inside of the horn.

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4. The loudspeaker of claim 1 wherein the side panels are curved in a direction corresponding to the diffraction slot opening while remaining straight in a direction perpendicular to the slot opening.

5. The loudspeaker of claim 1 wherein the first and second side panels have dimensions when flat corresponding to a section of a planar annulus defined by an inner radius, an outer radius, and an angle.

6. The loudspeaker of claim 1 wherein the first and second side panels each vary in length and curvature along the extent of the slot opening.

7. The loudspeaker of claim 1 wherein:

the manifold components each have two input openings and comprise two acoustic passages, each of the acoustic passages having a first end at a different one of the two input openings and a second end at the output opening, and the acoustic passages each curving away from the output opening in a different direction, such that the input openings are located near opposite sides of the horn.

8. The loudspeaker of claim 1 wherein:

the manifold components each have one input opening and comprise one acoustic passage having a first end at the input opening and a second end at the output opening, the acoustic passage of each manifold component curving away from the output opening in a direction opposite that of the neighboring manifold components' acoustic passages, such that the input opening is located near the opposite side of the horn from the neighboring manifold components' input openings.

9. A loudspeaker array comprising:

a first loudspeaker module comprising:

a first horn comprising a first end panel, a second end panel, a first side panel, and a second side panel, edges of at least the first and second side panels defining a first diffraction slot opening curved along a first radius,

a first plurality of manifold components, each comprising at least one acoustic passage and an output opening connected to the first diffraction slot opening, the output openings of the plurality of manifold components together constituting a diffraction slot source at the first diffraction slot opening; and

a first plurality of electro-acoustic transducers, each connected to an input opening of one of the manifold components of the first plurality; and

a second loudspeaker module comprising

a second horn comprising a third end panel, a fourth end panel, a third side panel, and a fourth side panel, edges of at least the third and fourth side panels defining a second diffraction slot opening curved along a second radius,

a second plurality of manifold components, each comprising at least one acoustic passage and an output opening connected to the second diffraction slot opening, the output openings of the plurality of manifold components together constituting a diffraction slot source at the second diffraction slot opening; and

a second plurality of electro-acoustic transducers, each connected to an input opening of one of the manifold components of the second plurality;

wherein the second end panel is adjacent to the third end panel, and the first diffraction slot opening and second diffraction slot opening together effectively form a continuous diffraction slot opening.

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10. A loudspeaker array comprising loudspeaker modules, each loudspeaker module comprising:

- a horn having a horizontal coverage angle and a vertical coverage angle, the horn comprising a top end panel, a bottom end panel, a first side panel, and a second side panel, edges of at least the first and second side panels defining a diffraction slot opening curved along a radius;
- a plurality of manifold components, each comprising at least one acoustic passage and an output opening connected to the diffraction slot opening, the output openings of the plurality of manifold components together constituting a diffraction slot source at the diffraction slot opening; and

a plurality of electro-acoustic transducers, each connected to an input opening of one of the manifold components, the array comprising:

- a first loudspeaker module having a first horizontal coverage angle, a first vertical coverage angle, and a first radius; and
- a second loudspeaker module having a second horizontal coverage angle greater than the first horizontal coverage angle, a second vertical coverage angle greater than the first vertical coverage angle, and a second radius less than the first radius,

wherein the bottom end panel of the first loudspeaker module is adjacent to the top end panel of the second loudspeaker module, and the first diffraction slot opening and second diffraction slot opening together effectively form a continuous diffraction slot opening.

11. The loudspeaker array of claim 10 further comprising: a third loudspeaker module having the first horizontal coverage angle, the first vertical coverage angle, and the first radius; and

a fourth loudspeaker module having a third horizontal coverage angle greater than the first and second horizontal coverage angles, a third vertical coverage angle greater than the first and second vertical coverage angles, and a third radius less than the first and second radii,

wherein the bottom end panel of the third loudspeaker module is adjacent to the top end panel of the first loudspeaker module, the top end panel of the fourth loudspeaker module is adjacent to the bottom end panel of

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the second loudspeaker module, and the first, second, third, and fourth diffraction slot openings together effectively form a continuous diffraction slot opening.

12. A method comprising:

obtaining loudspeaker array data identifying a number of loudspeaker modules to be used in a loudspeaker array, horizontal and vertical coverage dimensions for a listening area to be acoustically illuminated by the loudspeaker array, and a distance and angle from the loudspeaker array to the listening area;

computing a division of the listening area into a number of sections corresponding to the number of loudspeaker modules;

computing a horizontal and vertical coverage angle for each loudspeaker module such that each loudspeaker module illuminates a corresponding one of the divisions of the listening area based on the distance and angle from the loudspeaker array to the listening area, and automatically adjusting the computation of the horizontal and vertical coverage angles based on a desired radiation pattern of the loudspeaker array; and

outputting a list of the computed horizontal and vertical coverage angles for the loudspeaker modules.

13. The method of claim 12 wherein computing the horizontal and vertical coverage angle for each loudspeaker module comprises:

computing an ideal horizontal and vertical coverage angle; and

selecting a horizontal and vertical coverage angle pair from a pre-determined set of coverage angle pairs most closely matching the ideal horizontal and vertical coverage angles, wherein

for each loudspeaker module after the first loudspeaker module, computing the ideal vertical coverage angle includes adjusting the computation of the ideal vertical coverage angle to account for any discrepancy between the ideal vertical coverage angle computed for the previous loudspeaker module and the pre-determined vertical coverage angle selected for that previous loudspeaker module.

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